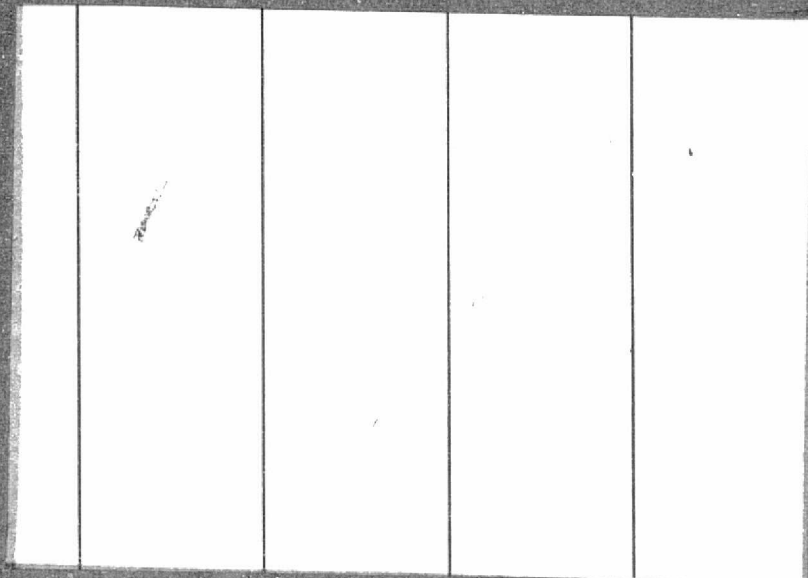


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UNIVERSITY OF WASHINGTON
DEPARTMENT OF MECHANICAL ENGINEERING



Sound and Vibration
Research Laboratory

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A STUDY OF SOUND
GENERATION IN SUBSONIC ROTORS

Report No. ME 73-11

Volume 2 of 2

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1.0 INTRODUCTION

Volume 2 of the final report is used to outline four computer programs which were developed during the grant period. All were coded in FORTRAN IV Language for usage on the University of Washington CDC 6400 digital computer.

The discussions here are primarily user input instructions. For two of the programs, ROTOR and AIRFOIL, the background models are developed in some detail in Volume 1 of the report. Program TONE is developed herein and SDATA is outlined here with reference to the literature where the detailed development requires a book to describe. The function of each program is discussed below.

- (1) Program AIRFOIL: This program computes the spectrum of radiated sound from a single airfoil immersed in a laminar flow field. The mechanism is force fluctuations which are related to the unsteady wake momentum. Input required is the aerodynamic properties of the turbulent wake.
- (2) Program ROTOR: The purpose of program ROTOR is to provide an extension of the single airfoil in AIRFOIL to a rotating frame. This is a model for sound generation in subsonic rotors. The only broad band mechanism currently in the program is the airfoil wake mechanism. Program ROTOR also computes tone sound generation due to the steady state forces on the blades (the Gutin result). Input requirements are similar to those for AIRFOIL.

- (3) Program TONE: A moving source analysis is used to generate a time series for an array of forces moving in a circular path as would be the case for a rotor. This analysis has great flexibility in that arbitrary blade spacings, blade forces and source motions can be easily accommodated. Broad band effects have been included using autoregression methods. The resultant time series are Fourier transformed using a Fast Fourier Transform to present the sound radiation in the more normal spectral form. With some study, a user can make this program a very valuable tool in sound generation investigations.
- (4) Program SDATA: This program is a standard time series analysis package. It will read in two discrete time series and form auto and cross covariances and normalize these to form correlations. The program will then transform the covariances to yield auto and cross power spectra by means of a Fourier transformation. The spectral data are presented in terms of amplitude and phase as well as the coherency spectrum.

A final note on these programs is in order. These were developed as tools to be used in the work performed under the grant. There was no intention to provide the programs as complete packages for general usage. The content is general enough, however, that with some effort a user should be able to employ the programs to advantage. He is cautioned to understand the analytical models before he becomes too involved in turning around decks. No program is a substitute for this understanding. Tools must be skillfully used.

To the end of providing a usable tool the programs are coded in variable names which read very like those of the physical variables they represent. For example the density will be RHO and the longitudinal turbulence covariance will be U1U1, etc. This should facilitate the reading of the program listings.

Each of the programs is discussed in the following sections.

2.0 PROGRAM AIRFOIL

The fortran program AIRFOIL is used to compute the sound radiated from a single airfoil due to those mechanisms related to the airfoil wake turbulence. A development of the model was presented in detail in Volume 1 of this report. Through the model, the sound radiation is related to the turbulent structure in the wake of the airfoil.

The equation which forms the basic result of the model and that programmed here is

$$G_p(\vec{x}, \omega) = \sum_{i=1}^2 \sum_{j=1}^2 16 S L_1^2(u_i u_j, \omega) L_2(u_i u_j, \omega) L_3(u_i u_j, \omega)$$

$$\omega^4 \int_{y_2} A_i A_j \overline{u_i u_j}(y_2, \omega) dy_2$$

with: $A_1 = \rho_0 U_1 \sin \psi / (4\pi a_0^3 |\vec{x}|)$

$$A_2 = \rho_0 \overline{U}_1 \cos \psi / (4\pi a_0^3 |\vec{x}|) .$$

Nomenclature used here is the same as that of Volume 1.

The integration noted above is performed using a Lagrangian method. This subroutine is part of the University of Washington Computer Library. Any simple numerical integration scheme may be substituted for this subroutine. In the program the subroutine is labeled LAGRAN.

2.1 Program Input

AIRFOIL is a simple computational program. It consists of three subprograms. The main program is AIRFOIL in which the actual computations are made. The next is subroutine DREAD in which data are read into and

conditioned to form the variables used in the actual computations. Finally, DWRITE in which the results of the computations are written out.

Input to the program is accomplished using IBM cards. In the following we will list each card type, the variable names and their descriptions.

CARD 1, FORMAT(7F10.2)

PATM	atmospheric static pressure	(atm)
TATM	" " temperature	(°C)
SPAN	airfoil length	(cm)
XLAM	length in the stream direction (this is a dead variable in the current program so any input is okay)	(cm)
DEL	wake thickness	(cm)
RAD	distance from airfoil to sound observer	(m)

CARD 2, FORMAT(2I5)

NPF	number of frequency points where data will be input
NPY	number of positions in the wake where data will be given

CARD 3, FORMAT(7F10.2)

Y(I) I = 1,NPY	positions in the wake measured relative to the wake centerline	(cm)
-------------------	----------------------------------------------------------------	------

Note that if more than 7 entries are needed, additional cards will be required for this variable.

CARD 4, FORMAT(7F10.2)

FREQ(I)	frequencies at which the following correlation lengths are to correspond	(Hz)
XL(I)	correlation length in stream direction for the longitudinal turbulent component	(cm)

YL(I)	correlation length across the wake for the longitudinal turbulence component	(cm)
ZL(I)	correlation length in the span direction for the longitudinal turbulence component	(cm)

CARD 5, FORMAT(7F10.2)

FREQ(I)	same as for CARD 4	
XT(I)	same as for CARD 4, but for the transverse turbulence component	(cm)
YT(I)	same as for CARD 4, but for the transverse turbulence component	(cm)
ZT(I)	same as for CARD 4, but for the transverse turbulence component	(cm)

Notice that CARD's 4 and 5 are read inside a DO loop. The statements are

```

      DO 16 I = 1,NPF
        READ(5,3) FREQ(I),XL(I),YL(I),ZL(I)
16    READ(5,3) FREQ(I),XT(I),YT(I),ZT(I)

```

Certainly more than just the two cards will be required here since more than one spectral point will generally be required. For this outline, we will continue with CARD 6 keeping in mind that CARD 6 refers to a new input type of data rather than the actual number of the card in the input file.

CARD 6, FORMAT(7F10.2)

XI(I)	ratio of wake velocity at Y(I) to the velocity in the free stream UMEAN	(--)
1 = 1,NPY		

CARD 7, FORMAT(7F10.2)

DDBI(I)	ratio of spectral level at FREQ(I) to the RMS level; this is for the longitudinal component of turbulence	(dB)
1 = 1,NPF		

DDB1 can be interpreted from the equation for its usage which is

$$u_1'(f) = u_1^2 * 10. **(-DDB1(I)/10.)$$

$u_1'(f)$ is the actual spectral level sought, and

u_1^2 is the RMS level for the longitudinal component of turbulence.

CARD 8, FORMAT(7F10.2)

DDB2(I)	the ratio of the spectral level at FREQ(I)	
I = 1,NPF	to the RMS level for the shear component	
	of turbulence	(dB)

see the note for DDB1

CARD 9, FORMAT(7F10.2)

DDB3(I)	the ratio of the spectral level at FREQ(I)	
	to the RMS level for the transverse component	
	of turbulence	(dB)

CARD 10, FORMAT(7F10.2)

U11(I)	spatial distribution of the RMS level of
I = 1,NPY	longitudinal turbulence across the wake,
	normalized with UMEAN

CARD 11, FORMAT(7F10.2)

U12(I)	spatial distribution of the RMS level of the
I = 1,NPY	shear stress across the wake, normalized
	with UMEAN

CARD 12, FORMAT(7F10.2)

U22(I)	spatial distribution of the RMS level of the
	transverse turbulence across the wake, normal-
	ized with UMEAN

This completes the input for the program.

Program output consists of a listing of the input variables and of the spectral distribution of the radiated sound pressure at each radiation angle 0° through 180° in 10° increments. The radiated sound is presented in dB by normalizing the pressure with a reference $20\mu\text{N}/\text{m}^2$.

2.2 Program Listing

```

PROGRAM AIRFOIL (INPUT,OUTPUT,TAPE5=INPUT,TAPE6=OUTPUT)

C   THIS PROGRAM COMPUTES THE SPECTRUM OF RADIATED SOUND FROM
C   CORRELATION AND SPECTRAL DATA GIVEN FOR TWO POINT VELOCITY
C   CORRELATIONS IN THE WAKE OF A SINGLE AIRFOIL

    DIMENSION XINTG(11),XINT1(11),XINT(11)
    DIMENSION FREQ(20),XL(20),YL(20),ZL(20),U1U1(11,20),X1(11),
    1U1U2(11,20),U2U2(11,20),Y(11),XM(11),XOMEG(20),
    2PSD(20),SPL(20),VCL(20),VCI(20),VAR(3),XI(20)
    DATA PI/3.14159/

C   READ IN DATA

C   PATM = STATIC PRESSURE (MM HG)
C   TATM = STATIC TEMPERATURE (DEG C)
CC   SPAN = AIRFOIL LENGTH (CM)
C   XLAM = STREAM INTEGRATION LENGTH (CM)
C   DEL = WAKE THICKNESS (CM)
C   UMEAN = FREE STREAM VELOCITY (M/SEC)
C   RAD = DISTANCE TO OBSERVER (M)
C   NPF = NUMBER OF FREQUENCY POINTS IN SPECTRUM
C   NPY = NUMBER OF SPATIAL POINTS IN WAKE
C   Y = LOCATIONS OF THE POINTS IN THE WAKE (CM)
C   FREQ = FREQUENCY ARRAY (HZ)
C   XL = STREAM CORRELATION FOR U1U1 (CM)
C   YL = CORRELATION ACROSS THE WAKE FOR U1U1 (CM)
C   ZL = SPAN CORRELATION FOR U1U1 (CM)
C   XT = STREAM CORRELATION FOR U2U2 (CM)
C   YT = CORRELATION ACROSS THE WAKE FOR U2U2 (CM)
C   ZT = SPAN CORRELATION FOR U2U2 (CM)
C   XI = RATIO OF WAKE VELOCITY TO UMEAN
C   DDB1 = RATIO OF SPECTRAL LEVEL TO RMS - U1U1 (DB)
C   DDB2 = RATIO OF SPECTRAL LEVEL TO RMS - U1U2 (DB)
C   DDB3 = RATIO OF SPECTRAL LEVEL TO RMS - U2U2 (DB)
C   U11 = RMS VALUE OF U1U1 (PERCENT)
C   U12 = RMS VALUE OF U1U2 (PERCENT)
C   U22 = RMS VALUE OF U2U2 (PERCENT)

    CALL DREAD(AO,RHO,PI,BM,XOMEG,VCL,VCI,SPAN,XLAM,DEL,XM,
    1RAD,U1U1,U1U2,U2U2,NPF,NPX,NPY,X,Y,FREQ,PATM,TATM,XL,XT)

C   COMPUTE RADIATION PATTERN

    COEF1=(RHO/(4.*PI*RAD))**2
    COEF2=COEF1*SPAN*BM**2
    DO 2000 KK=1,19
        UKK=KK
        CODE1=UKK
        THETA=-90.+10.*(UKK-1.)
        THET=THETA*PI/180.
        ST=SIN(THET)
        CT=COS(THET)
        STS=ST**2
        CTS=CT**2
        STCT=ST*CT

```

REPRODUCIBILITY OF THE
ORIGINAL PAGE IS POOR

COEF3=COEF2

C COMPUTE SPECTRUM

DO 1000 K=1,NPF

C COMPUTE INTEGRALS OVER THE WAKE VOLUME

XINT(K)=0.0

DO 300 II=1,3

DO 100 I=1,NPY

VAR(1)=ST3*U1U1(K,I)**2*VCL(K)*XL(K)

VAR(2)=STC1*U1U2(K,I)**2*(VCL(K)+VCT(K))/2.*(XL(K)+XT(K))/2.

VAR(3)=CT3*U2U2(K,I)**2*VCT(K)*XT(K)

100 XINTG(I)=VAR(II)

WRITE(6,10) (XINTG(I),I=1,NPY)

10 FORMAT(7E10.2)

CALL LAGRAN(Y,XINTG,NPY,1,1,0,ANS,IKKOR)

300 XINT(K)=ANS+XINT(K)

PSD(K)=XINT(K)*COFF3*XOMEG**4

PSD(K)=ABS(PSD(K))

WRITE(6,11) ANS

11 FORMAT(4H ANS 1E20.5)

1000 SPL(K)=10.*ALOG10(PSD(K)/400.E-12)

CALL DWRITE(AO,RHO,PI,BM,XOMEG,VCL,VCT,SPAN,XLAM,DEL,
1XM,KAD,U1U1,U1U2,U2U2,X,Y,THETA,PSD,SPL,NPY,NPX,NPF,
2FREQ,CODE1,PATM,TATM,XL,XT)

2000 CONTINUE

STOP

END

SUBROUTINE DREAD(AO,RHO,PI,BM,XOMEG,VCL,VCT,SPAN,XLAM,DEL,XM,
1RAD,U1U1,U1U2,U2U2,NPF,NPX,NPY,X,Y,FREQ,PATM,TATM,XL,XT)

C THIS SUBROUTINE READS AND CONDITIONS INPUT DATA

DIMENSION FREQ(20),XL(20),YL(20),ZL(20),U1U1(11,20),X1(11),
1U1U2(11,20),U2U2(11,20),Y(11),XM(11),XOMEG(20),

2 PSD(20),SPL(20),VCL(20),VCT(20)

DIMENSION DDB1(20),DDB2(20),DDB3(20),U11(11),U12(11),

1U22(11),DF1(20),DF2(20),DF3(20),X1(20),YT(20),Z1(20)

DATA CR,CU,GAM/288.,1.0,1.4/

READ(5,1) PATM,TATM,SPAN,XLAM,DEL,UMEAN,RAD

1 FORMAT(7F10.2)

CMFT = .0328

PATM = 2116.*PATM

TATM=(TATM+32.)*9./5.+460.

SPAN=SPAN*CMFT

XLAM=XLAM*CMFT

DEL=DEL*CMFT

UMEAN = UMEAN*CMFT*100.

RAD = RAD*CMFT*100.

```

      DSCALE=1.0
      AO=SQRT(GAM*CR*CG*TATM)
      RHO=PATM/(CG*CR*TATM)
      BM=UMEAN/AO
      READ(5,2) NPF,NPY
2  FORMAT(3I5)
      READ(5,1) (Y(I),I=1,NPY)
      DO 18 I=1,NPY
18  Y(I)=CMFT*Y(I)

      DO 16 I=1,NPF
      READ(5,3) FREQ(I),XL(I),YL(I),ZL(I)
16  READ(5,3) FREQ(I),XT(I),YT(I),ZT(I)

      DO 10 K=1,NPF
      XOMEG(K)=2.*PI*FREQ(K)
      FREQ(K)=FREQ(K)/DSCALE
      XOMEG(K)=XOMEG(K)/DSCALE
      COEF=2./(4.54*12.)
      XL(K)=XL(K)*COEF
      YL(K)=YL(K)*COEF
      ZL(K)=ZL(K)*COEF
      XT(K)=XT(K)*COEF
      YT(K)=YT(K)*COEF
      ZT(K)=ZT(K)*COEF
      VCL(K)=XL(K)*YL(K)*ZL(K)
10  VCT(K)=XT(K)*YT(K)*ZT(K)

      READ(5,1) (XI(I),I=1,NPY)

      DO 13 I=1,NPY
13  XM(I)=UMEAN*XI(I)/AO

      READ(5,1) (DDB1(K),K=1,NPF)
      READ(5,1) (DDB2(K),K=1,NPF)
      READ(5,1) (DDB3(K),K=1,NPF)
      DO 12 K=1,NPF
      DF1(K)=10.**(-DDB1(K)/20.)
      DF2(K)=10.**(-DDB2(K)/20.)
12  DF3(K)=10.**(-DDB3(K)/20.)
      READ(5,1) (U11(I),I=1,NPY)
      READ(5,1) (U12(I),I=1,NPY)
      READ(5,1) (U22(I),I=1,NPY)
      DO 15 K=1,NPF
      DO 15 I=1,NPY
      U1U1(K,I)=DF1(K)*U11(I)
      U1U2(K,I)=DF2(K)*U12(I)
15  U2U2(K,I)=DF3(K)*U22(I)

      3  FORMAT(4F10.2)
      RETURN
      END
      SUBROUTINE DWRITE(AO,RHO,PI,BM,XOMEG,VCL,VCT,SPAN,XLAM,DEL,XM,RAD,
101U1,U1U2,U2U2,X,Y,THETA,PSD,SPL,NPY,NPX,NPF,FREQ,CODE1,
2PATM,TATM,XL,XT)

```

```

    DIMENSION FREQ(20),XL(20),YL(20),ZL(20),U1U1(11,20),X1(11),
    1U1U2(11,20),U2U2(11,20),Y(11),XM(11),XOMEG(20),
    2 PSD(20),SPL(20),VCL(20),VCI(20)

```

```

    IF(CODE1.GT.1.) GO TO 21

```

```

    WRITE(6,1)

```

```

1  FORMAT(1H1,80H SOUND RADIATION FROM A SINGLE AIRFOIL AS RELATED TO
    1WAKE AERODYNAMIC PARAMETERS  //)

```

```

    WRITE(6,2) PATM,IATM,SPAN,XLAM,DEL,KAD

```

```

2  FORMAT(17H AMBIENT PRESSURE 1F10.2,20H AMBIENT TEMPERATURE
    11F10.2//20H DIMENSIONS OF INTEGRATION //13H AIRFOIL SPAN
    21E20.5,10X,17H STREAM DIRECTION 1E20.5/15H WAKE THICKNESS
    31E20.5//19H RADIUS TO OBSERVER 1E20.2//)

```

```

    WRITE(6,3) (Y(I),I=1,NPY)

```

```

3  FORMAT(30H STEADY FLOW MACH NUMBER FIELD//17H STREAM DIRECTION
    130X,16H ACROSS THE WAKE //20X,10F10.4/)

```

```

    WRITE(6,4) (XM(I),I=1,NPY)

```

```

4  FORMAT(5X,1F10.4,5X,10F10.5)

```

```

    DO 20 K=1,NPF

```

```

    WRITE(6,6) FREQ(K),VCL(K),VCI(K)

```

```

6  FORMAT(1H1,10H FREQUENCY 1E20.5,19H CORR VOLUME (LONG) 1E20.5,
    119H CORR VOLUME (TRAN) 1E20.5//)

```

```

    WRITE(6,7) (Y(I),I=1,NPY)

```

```

7  FORMAT(34H LONGITUDINAL TURBULENCE COMPONENT //17H STREAM DIRECTION
    1N 30X,10H ACROSS THE WAKE//20X,10F10.2/)

```

```

    WRITE(6,4) (U1U1(K,I),I=1,NPY)

```

```

    WRITE(6,8) (Y(I),I=1,NPY)

```

```

8  FORMAT(//32H TRANSVERSE TURBULENCE COMPONENT//17H STREAM DIRECTION
    130X,16H ACROSS THE WAKE //20X,10F10.2)

```

```

    WRITE(6,4) (U2U2(K,I),I=1,NPY)

```

```

    WRITE(6,9) (Y(I),I=1,NPY)

```

```

9  FORMAT(//47H SHEAR TURBULENCE COMPONENT //17H STREAM DIRECTION
    130X,16H ACROSS THE WAKE //20X,10F10.2/)

```

```

    WRITE(6,4) (U1U2(K,I),I=1,NPY)

```

```

20 CONTINUE

```

```

21 CONTINUE

```

```

    WRITE(6,11) THE1A

```

```

11 FORMAT(1H1,25H PREDICTED SOUND SPECTRUM //16H RADIATION ANGLE
    11F10.2//10H FREQUENCY 10X,17H RADIATED SPL(DB) )

```

```

    DO 12 I=1,NPF

```

```

12 WRITE(6,13) FREQ(I),SPL(I)

```

```

13 FORMAT(1F10.2,10X,1F10.2)

```

RETURN
END

2116.	530.	.292	.042	.013	209.	3.0
1.0						
6	5					
.08	-.04	0.0	0.4	0.8		
5000.	.6	.082	.324			
5000.	.49	.066	.084			
7500.	.6	.082	.324			
7500.	.56	.071	.094			
10000.	.6	.082	.273			
10000.	.62	.077	.120			
15000.	.43	.073	.189			
15000.	.58	.077	.159			
20000.	.42	.073	.157			
20000.	.58	.073	.137			
30000.	.28	.059	.105			
30000.	.45	.071	.105			
.92	.82	.54	.78	.84		
16.	15.5	15.5	12.	15.	19.	
20.	20.	20.	20.	20.	20.	
23.	21.	19.	10.	10.	17.	
.0055	.015	.04	.038	.007		
0.0	0.0	0.0	0.0	0.0	0.0	
.0055	.015	.038	.021	.005		

3.0 PROGRAM ROTOR

The purpose of developing program ROTOR was to provide a method for the prediction of sound generation from turbomachinery rotors. Aerodynamic operation was limited to the subsonic flow regime. Both tone noise and broad band effects were included. The tone noise part of the model was simply the steady state forces on the rotor blades. The broad band mechanism was taken as the force fluctuations which were relatable to the wake turbulence of the individual airfoils of the rotor. The mathematical model used for the basis of the sound radiation was that presented by Ffowcs Williams and Hawkings*.

To use the program one must supply the aerodynamic parameters for the particular design. Program ROTOR then computes the spectrum of sound radiated by the machine due to the mechanisms noted above.

The program is divided into five subroutines. A brief discussion of the contents of these subroutines will be given to provide some familiarization with the program structure. The names of the subroutines are: DATAIN, WAKE, ORTH1, TCNE, and ORTH2..

3.1 Subroutine DATAIN

As the name implies, DATAIN is used to read and condition the input data. The data input can be listed with the required FORMAT and physical units. The following are the only inputs required to run the program.

CARD 1, FORMAT(7F10.2)

RPM	rotor speed	(RPM)
R1	distance to observer	(M)

* "Theory Relating to the Noise of Rotating Machinery," J. Sound if Vib., 10, 1, 1969.

B	number of rotor blades	(-)
U1	blade relative velocity	(M/SEC)
XLAM	blade stagger angle	(DEGREES)
RS	source radius	(M)
FMAX	maximum frequency in spectrum	(HZ)

CARD 2, FORMAT(7F10.2)

FB	magnitude of force on each blade	(kg)
U1U1B	RMS turbulence level for the longitudinal component, normalized with U1	(--)
U1U2B	RMS shear turbulence, normalized with U1	(--)
U2U1B	RMS shear turbulence, normalized with U1	(--)
U2U2B	RMS turbulence level for the transverse component, normalized with U1	(--)

CARD 3, FORMAT(7F10.2)

ZL	blade span or other span length used to define the length of the source in the span direction	(M)
YL	thickness of the airfoil wake	(M)

CARD 4, FORMAT(1I5)

NF	number of frequency points to be used to input random data for wake related sound generation	(-)
----	----------------------------------------------------------------------------------------------	-----

In the remaining inputs a FORMAT(7F10.2) is used but if more than NF = 7 is used then more than one card will be needed for each of the input variables. Here we will list the variable arrays as if only one card will be required for the input.

ALL FOLLOWING CARDS, FORMAT(7F10.2)

F(I)	array of frequency values	(HZ)
CORR(I)	an arbitrary spectral shaping factor, normally set equal to 1	(--)
XLL(I)	correlation length in the stream direction, based upon the longitudinal component of turbulence	(CM)
YLL(I)	correlation length in the direction across the wake, based upon the longitudinal component of turbulence	(CM)
ZLL(I)	correlation length in the span direction, based upon the longitudinal component of turbulence	(CM)
XLT(I)	correlation length in the stream direction, based upon the transverse component of turbulence	(CM)
YLT(I)	correlation length in the direction across the wake, based upon the transverse component of turbulence	(CM)
ZLT(I)	correlation length in the span direction, based upon the transverse component of turbulence	(CM)

The following variables, U1U1, U1U2, U2U1, and U2U2, are spectral levels given in dB relative to the RMS levels for the component. The result of this procedure will be: if these turbulence quantities are given in dB down from RMS on a 1/3 octave spectrum then the radiated sound will also be 1/3 octave. If 6% spectra are used then the radiated sound will be 6%, also.

U1U1(I)	longitudinal component of turbulence	(dB)
U1U2(I)	shear stress component	(dB)
U2U1(I)	shear stress component	(dB)
U2U2(I)	transverse component of turbulence	(dB)

All input data are printed out in the physical units used in the computations.

3.2 Subroutine WAKE

This subroutine computes the wake related sound generation by the rotor. The model used to relate the force fluctuations on the rotor blading to aerodynamic variables is the single airfoil model. The expression which results and is here computed is

$$\left| p(\vec{x}, \omega) \right|^2 = \frac{1}{16\pi^2 r^2} \sum_{n=-\infty}^{\infty} \sum_{k=1}^B \left(\frac{\omega}{a_0} \right)^2 d_r(\theta_k, f-n\Omega) d_r^*(\theta_k, f-n\Omega) J_n^2 \left(\frac{-\omega R_s \sin \phi}{a_0} \right).$$

For the present case, the force fluctuations are

$$d_r(\theta_k, f-n\Omega) d_r^*(\theta_k, f-n\Omega) = S \delta_{\omega} (\omega - 2\pi n\Omega)^2 \rho_0^2 \sum_{\ell=1}^3 \sum_{m=1}^3 \hat{r}_{\ell} \hat{r}_m 16 L_1^2(u_{\ell}, u_m, \omega) L_2(u_{\ell}, u_m, \omega) L_3(u_{\ell}, u_m, \omega) \overline{u_{\ell} u_m}(\omega).$$

where: S = span, δ_w = wake thickness, L_i = correlation lengths ($i = 1, 3$), $\overline{u_{\ell} u_m}$ = spectral level of turbulent stress, ω = radian frequency, Ω = rotor speed, and f = frequency in Hz. J_n is a Bessel function of the first kind of integer order.

The output from this subroutine is the radiated sound pressure level expressed in dB relative to a reference pressure of $20 \mu\text{N/m}^2$.

3.3 Subroutine ORTH1

This subroutine computes the components of the position vector of the sound observer in the coordinate system relative to each of the rotor blades.

The product of these components with the force components on the blades yields the force in the direction of the observer. To do this the position vector must first be expressed in terms of the rotor coordinate system. It must then be rotated by the blade stagger angle to be in the blade coordinate system.

The observer position vector components for a typical blade now may be written:

$$r_1 = (r \sin \phi - R_s \cos \theta) / [r(1 - \frac{R_s}{r} \sin \phi \cos \theta)]$$

$$r_2 = -r_s \sin \theta / [r(1 - \frac{R_s}{r} \sin \phi \cos \theta)]$$

$$r_3 = \cos \phi / (1 - \frac{R_s}{r} \sin \phi \cos \theta)$$

The change from the observer coordinates to those of the rotor blade can be expressed in the rotation

$$\begin{pmatrix} -\sin \theta & \cos \theta & 0 \\ 0 & 0 & 1 \\ \cos \theta & \sin \theta & 0 \end{pmatrix}$$

and the further rotation by the stagger angle, λ ,

$$\begin{pmatrix} \sin \lambda & \cos \lambda & 0 \\ -\cos \lambda & \sin \lambda & 0 \\ 0 & 0 & 1 \end{pmatrix}$$

The stagger angle, λ , is the angle between the airfoil chord and a radial plane passing through the axis of the rotor.

3.4 Subroutine TONE

Subroutine TONE as currently set up computes the sound generation related to the steady state blade forces. This is the "Gutin" mechanism. The program has been developed for a rotating steady state stress term also but as yet it is not known how to model this in terms of aerodynamic design parameters.

The expression which is computed for the radiated sound pressure is

$$\left| p(\vec{x}, \omega) \right|^2 = \frac{1}{16\pi^2 r^2} \left(\frac{B}{R_s} \right)^2 \sum_{k=1}^B \sum_{m=-\infty}^{\infty} m^2 F_r^2(\theta_k, \omega) J_{mB}^2 \left(\frac{\omega R_s \sin \phi}{a_0} \right)$$

where: B = number of blades in the rotor

F_r = blade force in the direction of the observer

R_s = source of radius

ϕ = observer angle to rotor axis

r = distance to the observer

ω = radian frequency

3.5 Subroutine ORTH2

ORTH2 does exactly the same computation to the blade force as ORTH1 did for the random blade forces. The procedures are parallel.

```

PROGRAM ROTOR(INPUT,OUTPUT,TAPE5=INPUT,TAPE6=OUTPUT)

C   AERODYNAMIC SOUND GENERATION FROM MULTIBLADED ROTORS
C   ...THE RADIATED SOUND SPECTRUM IS PREDICTED BASED
C   ON STEADY LOADS AND AIRFOIL WAKE STATISTICAL PROPERTIES...

    DIMENSION U1U1(50),U1U2(50),U2U1(50),U2U2(50),F(50),FT(200),
    IRHOT(200),RHOS(30),CORR(30),XLL(30),XLT(30),VCL(30),VCT(30)

C   CALL THE SUBROUTINE TO INPUT AND CONDITION THE DATA

    CALL DATAIN(U1U1,U1U2,U2U1,U2U2,F,RPS,R1,B,U1,XLAM,RS,FMAX,FB,NF,
    ZL,YL,XLL,XLT,VCL,VCT,CORR)
    CALL WAKE(F,U1U1,U1U2,U2U1,U2U2,NF,RPS,FMAX,RS,RHOS,B,
    IR1,XLAM,U1,ZL,YL,XLL,XLT,VCL,VCT,CORR)

C   WAKE COMPUTES BROAD BAND SOUND GENERATION

    CALL TONE(R1,B,FB,IRHOT,XLAM,RPS,RS,U1,FT)
C   TONE COMPUTES LEVELS OF PURE TONE GENERATED BY ROTATING
C   STEADY BLADE FORCE AND MEAN VALUES OF STRESS TENSOR
    STOP
    END
    SUBROUTINE DATAIN(U1U1,U1U2,U2U1,U2U2,F,RPS,R1,B,U1,XLAM,RS,
    IFMAX,FB,NF,ZL,YL,XLL,XLT,VCL,VCT,CORR)

C   READS AND CONDITIONS DATA

    DIMENSION U1U1(50),U1U2(50),U2U1(50),U2U2(50),F(50)
    DIMENSION VC(30),CORR(30),XLL(30),YLL(30),ZLL(30),
    ZLXT(30),YLT(30),ZLT(30),VCL(30),VCL(30)

C   RPM = ROTOR SPEED (RPM)
C   R1 = OBSERVER RADIUS (M)
C   B = NUMBER OF ROTOR BLADES
C   U1 = BLADE RELATIVE VELOCITY (M/SEC)
C   XLAM = STAGGER ANGLE - MEASURED FROM FAN AXIS (DEG)
C   RS = SOURCE RADIUS (M)
C   FMAX = MAXIMUM FREQUENCY OF INTEREST (HZ)
C   FB = BLADE FORCE (KG)
C   U1U1B = RMS LONGITUDINAL TURBULENCE - NORMALIZED WITH U1
C   U1U2B = RMS SHEAR - NORMALIZED WITH U1
C   U2U1B = RMS SHEAR - NORMALIZED WITH U1
C   U2U2B = RMS TRANSVERSE TURBULENCE - NORMALIZED WITH U1
C   ZL = SPAN DIMENSION (CM)
C   ZL = WAKE THICKNESS (CM)
C   CHD = BLADE CHORD (CM)
C   NF = NUMBER OF FREQUENCY POINTS
C   F(I) = FREQUENCY ARRAY (HZ)
C   CORR(I) = A SPECTRAL WEIGHTING FUNCTION (NORMALLY UNITY)
C   XLL(I) = STREAM CORRELATION LENGTH FOR U1U1B
C   YLL(I) = NORMAL CORRELATION LENGTH FOR U1U1B (CM)
C   ZLL(I) = SPAN CORRELATION LENGTH FOR U1U1B (CM)
C   ZLT(I) = STREAM CORRELATION LENGTH FOR U2U2B (CM)
C   YLT(I) = NORMAL CORRELATION LENGTH FOR U2U2B (CM)
C   ZLT(I) = SPAN CORRELATION LENGTH FOR U2U2B (CM)

```

```

C      U1U1(I) = DB DOWN FROM U1U1B (RMS)
C      U1U2(I) = DB DOWN FROM U1U2B (RMS)
C      U2U1(I) = DB DOWN FROM U2U1B (RMS)
C      U2U2(I) = DB DOWN FROM U2U2B (RMS)

```

```
DATA AO,CG,PI/342.,1.0,3.14159/
```

```

READ(5,1) RPM,R1,B,U1,XLAM,RS,FMAX
READ(5,1) FB,U1U1B,U1U2B,U2U1B,U2U2B
READ(5,1) ZL,YL,XL,CHD

```

```
CMFT = .0328
```

```
R1=R1*CMFT*100.
```

```
U1=U1*CMFT*100.
```

```
RS=RS*CMFT*100.
```

```
FB=FB*2.21
```

```
ZL=ZL*CMFT
```

```
YL=YL*CMFT
```

```
CHD=CHD*CMFT
```

```
1 FORMAT(7F10.2)
```

```
TPI=2.*PI
```

```
XLAM=XLAM*PI/180.
```

```
RPS=TPI*RPM/60.
```

```
READ(5,2) NF
```

```
2 FORMAT(11I5)
```

```
READ(5,1) (F(I),I=1,NF)
```

```
READ(5,1) (CORR(I),I=1,NF)
```

```
READ(5,1) (XLL(I),I=1,NF)
```

```
READ(5,1) (YLL(I),I=1,NF)
```

```
READ(5,1) (ZLL(I),I=1,NF)
```

```
READ(5,1) (XLT(I),I=1,NF)
```

```
READ(5,1) (YLT(I),I=1,NF)
```

```
READ(5,1) (ZLT(I),I=1,NF)
```

```
READ(5,1) (U1U1(I),I=1,NF)
```

```
READ(5,1) (U1U2(I),I=1,NF)
```

```
READ(5,1) (U2U1(I),I=1,NF)
```

```
READ(5,1) (U2U2(I),I=1,NF)
```

```
C      INTRODUCE SCALING FACTORS
```

```
SCL=(CHD/3.0)**0.8*(U1/150.)**(-0.2)
```

```
SCF=(U1/150.)**1.2*(CHD/3.0)**(-0.8)
```

```
FMAX=FMAX*SCF
```

```
XL=XL*SCL
```

```
YL=YL*SCL
```

```
ZL=ZL*SCL
```

```
DO 30 I=1,NF
```

```
F(I)=F(I)*SCF
```

```
XLL(I)=XLL(I)*SCL
```

```
YLL(I)=YLL(I)*SCL
```

```
ZLL(I)=ZLL(I)*SCL
```

```
XLT(I)=XLT(I)*SCL
```

```
YLT(I)=YLT(I)*SCL
```

```
30 ZLT(I)=ZLT(I)*SCL
```

```
DO 20 I=1,NF
```

```
VCL(I)=8.*XLL(I)*YLL(I)*ZLL(I)/28350.
```

```
20 VCI(I)=8.*XLT(I)*YLT(I)*ZLT(I)/28350.
```


C WRITE THE INPUT DATA

```

WRITE(6,3) R1, RPM, FB, XLAM, B, RS, U1, FMAX, U1U1B, U1U2B, U2U1B, U2U2B
3 FORMAT(1H1, 54H AERO DYNAMIC SOUND GENERATION FROM MULTIBLADED ROTOR
13//23H OBSERVER DISTANCE (M) 1F7.2, 10X, 12H ROTOR SPEED
21F10.2/19H STEADY BLADE FORCE 1F11.2, 10X, 20H BLADE STAGGER ANGLE
31F10.2/17H NUMBER OF BLADES 1F3.0, 20X, 14H SOURCE RADII
41F0.2/24H BLADE RELATIVE VELOCITY 1F0.1, 10X, 10H MAX. FREQUENCY
51F10.2/10H U1U1(RMS) 1E10.3, 5X, 10H U1U2(RMS) 1E10.3, 5X,
610H U2U1(RMS) 1E10.3, 5X, 10H U2U2(RMS) 1E10.3//)
WRITE(6,9) ZL, YL
9 FORMAT(21H WAKE VOLUME L(Z) = 1E15.5, 5X,
18H L(Y) = 1E15.5//)
WRITE(6,11)
11 FORMAT(100H FREQUENCY L(Z) L(Y)
1 L(X) SCALAR FACTOR )
WRITE(6,12) (F(I), ZLL(I), YLL(I), XLL(I), CORR(I), I=1, NF)
WRITE(6,14) (F(I), ZLI(I), YLI(I), XLI(I), CORR(I), I=1, NF)
12 FORMAT(5E20.5)
WRITE(6,4)
4 FORMAT(14H SPECTRUM U1U1 //)
WRITE(6,5) (F(I), U1U1(I), I=1, NF)
5 FORMAT(4(1F10.3, 1E20.4))
WRITE(6,6)
6 FORMAT(///14H SPECTRUM U1U2 //)
WRITE(6,5) (F(I), U1U2(I), I=1, NF)
WRITE(6,7)
7 FORMAT(///14H SPECTRUM U2U1 //)
WRITE(6,5) (F(I), U2U1(I), I=1, NF)
WRITE(6,8)
8 FORMAT(///14H SPECTRUM U2U2 //)
WRITE(6,5) (F(I), U2U2(I), I=1, NF)

```

C CONDITION TURBULENCE DATA TO LINEAR SCALE

```

DO 10 I=1, NF
XLL(I)=XLL(I)/(2.54*12.)
XLI(I)=XLI(I)/(2.54*12.)
U1U1(I)=U1U1B/(10.**((U1U1(I)/20.))
U1U2(I)=U1U2B/(10.**((U1U2(I)/20.))
U2U1(I)=U2U1B/(10.**((U2U1(I)/20.))
10 U2U2(I)=U2U2B/(10.**((U2U2(I)/20.))
RETURN
END
SUBROUTINE WAKE(F, U1U1, U1U2, U2U1, U2U2, NF, RPM, FMAX, RS, RHOS, B,
R1, XLAM, U1, ZL, YL, XLL, XLI, VCL, VCI, CORR)

```

C COMPUTES SOUND RADIATION FROM WAKE TURBULENCE PROPERTIES

```

DIMENSION F(30), U1U1(30), U1U2(30), U2U1(30), U2U2(30), BES(1000),
IBEN(1000), PHI(20), SX(30,3), S1(30),
2RHOS(30), V1(3), VC(30), CORR(30),
3XLL(30), XLI(30), VCL(30), VCI(30)

```

```
DATA CG, AO, PI, RHO/1.0, 342., 3.14159, 1.175/

```

```

PREF=20.E-06/AO**2
PREF=PREF*PREF
C1=1./(4.*PI*AO**2*R1)**2
TPI=2.*PI
C2=C1*(RHO/CG)**2
REV=RPS/TP1
ATOT=ZL*YL

```

C COMPUTE THE RADIATION PATTERN

```

DO 1000 KK=1,3
PHI(KK)=45.*FLOAT(KK-1)
PHIP=PHI(KK)*PI/180.

```

C COMPUTE THE SPECTRUM

```

DO 11 I=1,NF
DVL=VCL(I)*ATOT*CORK(I)
DVT=VCT(I)*ATOT*CORK(I)
DVL=SQRT(DVL)
DVT=SQRT(DVT)
DV=0.0
SX(I,1)=-01U1(I)*DVL*SQRT(XLL(I))
SX(I,2)=02U2(I)*DVT*SQRT(XLL(I))
DO 40 J=1,3
40 WRITE(6,41) SX(I,J)
41 FORMAT(10H SX(I,J) = 1E20.5)
DO 32 J=1,3
32 V1(J)=SX(I,J)
V3=01U2(I)
V4=02U1(I)
FRA=F(I)*PI

CALL ORTH+(V1,FRA,V3,V4,PHIP,ALAM,B,R1,R3,DD)

11 S1(I)=DD
DO 2000 K=1,NF
C3=(TPI*F(K)/AO)**2*U1**2

```

C THE ARGUMENT FOR THE BESSEL FUNCTION

```

ARG=-TPI*F(K)*R3*SIN(PHIP)/AO

```

C NMAX IS THE MAXIMUM NECESSARY VALUES OF N FOR THE BESSEL FUNCTION

```

NMAX=IFIX(FMAX*TP1/RPS)*2+1
NMAX=250
IF(NMAX.GT.400) NMAX=400
IF(ARG.NE.0.0) GO TO 60
BES(1)=1.0
DO 61 LL=2,NMAX
61 BES(LL)=0.0
GO TO 62
60 CONTINUE
DO 63 LL=1,NMAX

```

REPRODUCIBILITY OF THE
ORIGINAL PAGE IS POOR

```

63 BES(LL)=0.0
   CALL NBESJ(ARG,NMAX,BES)
62 CONTINUE

```

C SQUARE THE BESSEL FUNCTIONS

```

DO 44 J=1,NMAX
  BEN(J)=(-1.)*BES(J)
  BES(J)=BES(J)*BES(J)
44 BEN(J)=BEN(J)*BEN(J)

```

C PROVIDE THE SUMMANDS

```

SUM1=0.0
DO 14 N=1,NMAX
  FP=F(K)+FLOAT(N-1)*RLV
  IF(FP.GT.FMAX) GO TO 13
  DUM1=1BLU1(FP,F,3,1,NF)
  DUM1=DUM1*(TPI*FP)**2
  GO TO 14
13 DUM1=0.0
14 SUM1=SUM1+DUM1*BES(N)
  DO 12 N=2,NMAX
    FN=F(K)-FLOAT(N-1)*RLV
    IF(FN.LT.-FMAX) GO TO 15
    FN=ABS(FN)
    DUM1=1BLU1(FN,F,3,1,NF)
    DUM1=DUM1*(TPI*FN)**2
    GO TO 12
15 DUM1=0.0
12 SUM1=SUM1+DUM1*BEN(N)
  WRITE(6,31) SUM1,DUM1
31 FORMAT(5H SUM1 1E20.5,5H DUM1 1E20.5)

```

C THIS COMPLETES THE DIPOLE TERM

C WRITE THE MODULUS OF THE RADIATED SOUND DENSITY

```

VAR1=C2*C3*SUM1/PREF
WRITE(6,50) F(K),VAR1
50 FORMAT(1F10.3,1E20.5)
2000 RHOS(K)=C2*C3*SUM1
  WRITE(6,1) PHI(KK)
  1 FORMAT(1H1,37H SPECTRAL DISTRIBUTION OF ROTOR SOUND ///
    130H WAKE TURBULENCE RELATED SOUND ///17H RADIATION ANGLE 1F7.2//
    210X,10H FREQUENCY 10X,8H SPL(DB) /)
  DO 10 I=1,NF
    VAR1=RHOS(I)/PREF
    IF(VAR1.LT.1.0) GO TO 10
    RHOS(I)=10.*ALOG10(VAR1)
  10 CONTINUE
  WRITE(6,2) (F(I),RHOS(I),I=1,NF)
  2 FORMAT(12X,1F10.2,10X,1E20.4)
1000 CONTINUE
  RETURN
  END

```

```
SUBROUTINE ORTH1(SX,KPS,U102,U201,PHI,XLAM,B,R1,R3,DRDR)
```

```
C COMPUTES THE SUM OF THE COMPONENTS IN THE DIRECTION OF THE  
C OBSERVER FROM EACH BLADE OF THE ROTOR
```

```
DIMENSION ZY(3,3),THE(30),YX(3,3),SX(3),SYY(3,3),K(3),V(3)
```

```
C INTRODUCE THE COMPONENTS OF THE FIRST ROTATION
```

```
PI=3.14159  
ZY(1,1)=SIN(XLAM)  
ZY(1,2)=COS(XLAM)  
ZY(1,3)=0.0  
ZY(2,1)=-COS(XLAM)  
ZY(2,2)=SIN(XLAM)  
ZY(2,3)=0.0  
ZY(3,1)=0.0  
ZY(3,2)=0.0  
ZY(3,3)=1.0
```

```
C THE NEXT ROTATION DEPENDS ON THE INITIAL BLADE POSITION AS WELL  
C THE OBSERVER RADII VECTOR
```

```
NB=IFIX(B)  
DRDR=0.0  
DO 10 K=1,NB  
THE(K)=360./B*FLOAT(K)  
THET=THE(K)*PI/180.  
DEN=R1*(1.-R3/R1*SIN(PHI)*COS(THET))  
R(1)=(R1*SIN(PHI)-R3*COS(THET))/DEN  
R(2)=(-R3*SIN(THET))/DEN  
R(3)=(R1*COS(PHI))/DEN  
YX(1,1)=-SIN(THET)  
YX(1,2)=COS(THET)  
YX(1,3)=0.0  
YX(2,1)=0.0  
YX(2,2)=0.0  
YX(2,3)=1.0  
YX(3,1)=COS(THET)  
YX(3,2)=SIN(THET)  
YX(3,3)=0.0
```

```
C NOW SUM ON THE COMPONENTS TO GET FINAL VALUES IN OBSERVER FRAME
```

```
DD=0.0  
DO 11 I=1,3  
V(I)=0.0  
DO 11 J=1,3  
DO 11 L=1,3  
11 V(I)=R(L)*ZY(I,J)*YX(J,L)+V(I)  
WRITE(6,30) (V(I),I=1,3)  
30 FORMAT(5H V(B) 3E15.3)
```

```
C DETERMINE THE COMPONENTS IN THE DIRECTION OF THE OBSERVER
```

```
C DIPOLE TERMS
```

```

      DO 12 I=1,2
      DO 12 J=1,2
      IF(I.EQ.1.AND.J.EQ.2) GO TO 13
      IF(J.EQ.1.AND.I.EQ.2) GO TO 131
      VAR1=SX(I)*SX(J)
      GO TO 12
13  VAR1=-U1U2**2
      GO TO 12
131 VAR1=-U2U1**2
      12 DD=DD+V(I)*VAR1*V(J)
      10 DRDR=DRDR+DD
      WRITE(6,20) DRDR
      20 FORMAT(1E20.5)

C      THIS COMPLETES COMPUTATION OF THE COMPONENTS IN THE
C      DIRECTION OF THE OBSERVER

      RETURN
      END
      SUBROUTINE TONE(R1,B,FB,KHU,XLAM,RPS,RS,U1,F)

C      COMPUTES PURE TONES GENERATED BY STEADY BLADE FORCES AND
C      THE MEAN VALUES OF THE STRESS TENSOR

      DIMENSION BES(200),SX(3),SYY(3,3),IX(200),IYY(200),F(200),
      IRHU(200),PHI(20),TT(30),DD(30)
      DATA AO,PI,CG/342.,3.14159,1.0/
      PREF=20.E-06/AO**2
      PREF=PREF*PREF
      RATM=1.175
      RATM=0.0
      NB=IFIX(B)

C      PROVIDE THE TENSOR AND FORCE COMPONENTS

      C1=1./(2.*PI*AO**2*R1)**2
      REV=RPS/(2.*PI)
      FAX=0.05*FB
      FT=0.95*FB
      DO 1 I=1,3
1  SX(I)=0.0
      SX(1)=-FAX
      SX(2)=FT
      DO 2 I=1,3
      DO 2 J=1,3
2  SYY(I,J)=0.0
      SYY(1,1)=1.0
      DO 1000 KK=1,1
      PHI(KK)=30.+10.*FLOAT(KK)
      PHI(KK)=60.+10.*FLOAT(KK)
      PHI(KK)=80.
      PHIP=PHI(KK)*PI/180.
      C2=(B/RS)**2
      C3=(B*RATM*U1*U1/(AO*RS*SIN(PHIP)))**2

```

```

C      COMPUTE THE SPECTRUM
      CALL ORTH2(SX,SY,PHIP,XLAM,B,R1,RS,DD,11)

C      COMPUTE 5 HARMONICS OF BLADE PASSAGE FREQUENCY
      NH=10
      DO 10 N=1,NH
      F(N)=FLOAT(N)*REV*B
      ARG=2.*PI*F(N)*RS*SIN(PHIP)/AO
      NN=IFIX(B)*N+1
      CALL NBESJ(ARG,NN,BES)
      WRITE(6,30) ARG,BES(NN)
30  FORMAT(16H BESSEL FUNCTION      2E20.5)
      TYY(N)=0.0
      TX(N)=0.0
      DO 40 K=1,NB
      DO 40 L=1,NB
      TX(N)=TX(N)+BES(NN)*BES(NN)*DD(K)*DD(L)
40  TYY(N)=TYY(N)+BES(NN)*BES(NN)*T1(K)*T1(L)*2.*PI*F(N)

C      THIS COMPLETES THE SPECTRAL LEVEL COMPUTATION - NOW CALL ROUTINE
C      TO PERFORM COORDINATE TRANSFORMATIONS

      VAR1=C1*C2*TX(N)/PREF
      VAR2=C1*C3*TYY(N)/PREF
      WRITE(6,20) F(N),VAR1,VAR2
20  FORMAT(3(1F10.2,2F15.3))
      FNN=FLOAT(N)*FLOAT(N)
10  RHO(N)=C1*C2*FNN*TX(N)
      WRITE(6,5)
5   FORMAT(1H1,37H SPECTRAL DISTRIBUTION OF ROTOR SOUND      ///)
      DO 11 I=1,NH
      VAR2=RHO(I)/PREF
      IF(VAR2.LT.1.0) GO TO 11
      RHO(I)=10.*ALOG10(VAR2)
11  CONTINUE
      WRITE(6,3) PHI(KK)
3   FORMAT(/////36H SPECTRAL DISTRIBUTION OF TONE SOUND      ///
117H RADIATION ANGLE  1F5.2///
29H FREQ(HZ)      11X,8H SPL(DB)      //)
      WRITE(6,4) (F(I),RHO(I),I=1,NH)
4   FORMAT(4(1F10.2,1E20.3))
1000 CONTINUE
      RETURN
      END
      SUBROUTINE ORTH2(SX,SY,PHI,XLAM,B,R1,RS,DRDR,TRTR)

C      COMPUTES THE SUM OF THE COMPONENTS IN THE DIRECTION OF THE
C      OBSERVER FROM EACH BLADE OF THE ROTOR

      DIMENSION ZY(3,3),THE(30),YX(3,3),SX(3),SY(3,3),R(3),V(3)
      DIMENSION DRDR(30),TRTR(30)

C      INTRODUCE THE COMPONENTS OF THE FIRST ROTATION

```

```

PI=3.14159
ZY(1,1)=SIN(XLAM)
ZY(1,2)=COS(XLAM)
ZY(1,3)=0.0
ZY(2,1)=-COS(XLAM)
ZY(2,2)=SIN(XLAM)
ZY(2,3)=0.0
ZY(3,1)=0.0
ZY(3,2)=0.0
ZY(3,3)=1.0

```

```

C   THE NEXT ROTATION DEPENDS ON THE INITIAL BLADE POSISTION AS DOES
C   THE OBSERVER RADIUS VECTOR

```

```

NB=IFIX(B)
DO 10 K=1,NB
THE(K)=360./B*FLOAT(K)
THET=THE(K)*PI/180.
DEN=R1*(1.-RS/R1*SIN(PHI)*COS(THET))
R(1)=(R1*SIN(PHI)-RS*COS(THET))/DEN
R(2)=(-RS*SIN(THET))/DEN
R(3)=(R1*COS(PHI))/DEN
YX(1,1)=-SIN(THET)
YX(1,2)=COS(THET)
YX(1,3)=0.0
YX(2,1)=0.0
YX(2,2)=0.0
YX(2,3)=1.0
YX(3,1)=COS(THET)
YX(3,2)=SIN(THET)
YX(3,3)=0.0

```

```

C   NOS SUM ON THE COMPONENTS TO GET FINAL VALUES IN OBSERVER FRAME

```

```

DD=0.0
TT=0.0
DO 11 I=1,3
V(I)=0.0
DO 11 J=1,3
DO 11 L=1,3
11 V(I)=R(L)*YX(J,L)*ZY(I,J)+V(I)
WRITE(6,20) (V(I),I=1,3)
20 FORMAT(/31H OBSERVER VECTOR IN BLADE BASIS /3E20.5/)
C   DETERMINE THE COMPONENTS IN THE DIRECTION OF THE OBSERVER
C   DIPOLE TERMS
DO 12 I=1,2
12 DD=DD+V(I)*SX(I)
WRITE(6,30) DD
30 FORMAT(16H FORCE COMPONENT 1E20.5)
C   QUADRAPLES
DO 18 I=1,2
DO 18 J=1,2
18 TT=TT+V(I)*SYY(I,J)*V(J)
DRDR(K)=DD
10 TRTR(K)=TT
C   THIS COMPLETES COMPUTATION OF THE COMPONENTS IN THE
C   DIRECTION OF THE OBSERVER
RETURN
END

```

4.0 PROGRAM TONE

This model was set up in a computer program so that direct numerical computations of the time series described by the equation for the radiated sound could be made. This direct approach of calculating the actual time series is the difference between the present approach and those of other rotor models. Typically the analytical results for the rotor are Fourier transformed and reduced to a closed form (such as program ROTOR described herein). This approach normally requires that simplifications be made so that the closed form can be obtained. The advantage of the present approach is no simplifications in the analysis need be made.

The time series is computed for a sufficient number of values so that a good statistical sample will result. The time series is then Fourier transformed using a Fast Fourier Transform algorithm to yield the spectrum of the radiated sound. In other words, the model is used to generate a time series which is handled exactly as one would handle experimental data. Very complex source histories can be studied with no more effort than the very simple steady force case. For example, rotor wobble can be simulated in order to study how it affects the generation of sound; the individual blades can be allowed to vibrate as might occur in flutter cases; or each blade can be assigned a different value of force to determine the effect of such blade to blade differences as might be caused by manufacturing imperfections.

Of most importance, is that complex random effects can be included in this model, also. Such mechanisms as force fluctuations related to wake turbulence, inflow turbulence, or even rotating stalls can be studied. The

way this must be done is to express the force fluctuation in a time series that has the required statistical nature. If the spectrum of the force fluctuation relative to the blade is known, then it is possible to generate a random time series which will give the same statistics as the known result. The methods for generating an auto-regression analysis can be used to simulate the spectrum of the force on the blades. Once the time series are known, the results from the different blades may be combined in different ways to simulate different mechanisms of sound generation. For example, the effect of large scale turbulence can be simulated by allowing several of the blades to experience the same force fluctuation, or allow all of them to experience it. The case where there is no blade to blade correlation would model the wake related sound generation.

The idea of the model is to provide a tool where the various mechanisms which might produce appreciable sound generation can be evaluated. In addition, it is sought to set up the model in such a way that its usage will be reasonably simple and straightforward. As an example, if one wants to study the effect of different forces on each of the blades, he merely reads in the array of steady loads he wishes the rotor to have, and that is all. Rotor wobble is accounted for by replacing the velocity of the source, the rotor velocity, with a time dependent function describing the wobble characteristics.

4.1 Tone Noise Generation Model

The generation of pure tone sound due to the steady state forces on the rotor blading is studied in the model. In addition, an auto-regression method is used to write a time series which approximates a random force distribution resulting in broad band noise generation.

The model is based upon M.V. Lowson's development for rotating point forces^{*}. The development is done for a force per unit volume. In this representation only the point force case will be considered. Extension to the distributed case can be treated as an array of point forces where the array is distributed along the span as well as circumferentially. This has been done by the author and the mechanics are straightforward but the expense in computer time is correspondingly increased.

For a point force, the radiated sound in the far field can be written:

$$p(\vec{x}, t) = \left[\frac{x_i - y_i}{4\pi a_0 r^2 (1-M_r)^2} \left\{ \frac{\partial F_i}{\partial t} + \frac{F_i}{1-M_r} \frac{\partial M_r}{\partial t} \right\} \right]_{t_R}$$

where the evaluation of the source terms of the retarded time

$$t_R = t - \frac{|\vec{x} - \vec{y}|}{a_0}$$

t_R = time of sound generation

t = time of observation, and $|\vec{x} - \vec{y}|/a_0$ = propagation time.

The position vector $\vec{r} = \vec{x} - \vec{y}$ can be written in the form

$$\vec{r} = \sum_{i=1}^3 (x_i - y_i) \vec{e}_i \text{ in rectangular cartesian coordinates.}$$

The Mach number M_r is the component of the source Mach number in the direction of the observer. It may be written

$$M_r = \vec{M}_s \cdot \vec{r} / |\vec{r}|$$

or in components as

* "The Sound Field for Singularities in Motion," Proc. Roy. Soc., A, 236, 559, 1965.

$$M_r = \sum_{i=1}^3 M_i (x_i - y_i) / |\vec{r}|.$$

The components of the blade force F can be chosen to represent the thrust and torque loads on the individual blades. Using the blade stagger angle, ξ , write

$$F_1 = -F \cos \xi$$

$$F_2 = -F \sin \xi \sin \theta$$

$$F_3 = F \sin \xi \cos \theta.$$

Take the time derivatives of these force components:

$$\partial F_1 / \partial t = 0$$

$$\partial F_2 / \partial t = -F\Omega \sin \xi \cos \theta$$

$$\partial F_3 / \partial t = -F\Omega \sin \xi \sin \theta.$$

The components of the position of the observer are written in polar form as

$$x_1 = r \cos \phi$$

$$x_2 = r \sin \phi$$

$$x_3 = 0.$$

The location of the source is also written in polar form:

$$y_1 = 0$$

$$y_2 = R_s \cos \theta$$

$$y_3 = R_s \sin \theta.$$

Finally, the components of the source Mach number are

$$M_1 = 0$$

$$M_2 = -M \sin \theta$$

$$M_3 = M \cos \theta.$$

The value of the source Mach number is

$$M_s = R_s \Omega / a_o.$$

The Mach number component M_r may now be computed:

$$M_r = \sum_{i=1}^3 M_i (x_i - y_i) / r.$$

Substitution yields

$$\begin{aligned} M_r &= (r_1 \cos \phi)(0)/r + (r_1 \sin \phi - R_s \cos \theta)(-M_s \sin \theta)/r \\ &= -(r_1/r) M_s \sin \phi \sin \theta + (R_s/r) M_s \cos \theta \sin \theta, \end{aligned}$$

so that to the first order

$$M_r = -M_s (r_1/r) \sin \phi \sin \theta.$$

The time derivative of the Mach number becomes

$$\partial M_r / \partial t = -M_s (r_1/r) \Omega \sin \phi \cos \theta.$$

All the terms required for the computation of the pressure radiated are now complete. The radiated sound is

$$\begin{aligned}
p(\vec{x}, t) = & \left[\frac{F}{4\pi a_0 r^2 (1-M_r)^2} \left\{ \frac{r_1 \sin \phi - R_s \cos \theta}{r} (-\Omega \sin \xi \cos \theta) \right. \right. \\
& \frac{-R_s \sin \theta}{r} (-\Omega \sin \xi \sin \theta) + \frac{1}{1-M_r} \frac{\partial M_r}{\partial t} \left[\frac{r_1 \cos \phi}{r} (-\cos \xi) \right. \\
& \left. \left. + \frac{r_1 \sin \phi - R_s \cos \theta}{r} (-\sin \xi \sin \theta) + \frac{-R_s \sin \theta}{r} \sin \xi \cos \theta \right] \right\} \right].
\end{aligned}$$

And, expanding:

$$\begin{aligned}
p(\vec{x}, t) = & \left[\frac{F}{4\pi a_0 r (1-M_r)^2} \frac{-r_1 \Omega \sin \phi \sin \xi \cos \theta}{r} \right. \\
& + \frac{R_s \Omega \sin \xi}{r} (\cos^2 \theta + \sin^2 \theta) + \frac{1}{1-M_r} \frac{\partial M_r}{\partial t} \left[\frac{-r_1 \cos \phi \cos \xi}{r} \right. \\
& \left. \left. \frac{-r_1 \sin \xi \sin \phi \sin \theta}{r} \right] \right].
\end{aligned}$$

The solution reduces to the form

$$\begin{aligned}
p(\vec{x}, t) = & \frac{F}{4\pi a_0 r^2 (1-M_r)^2} [\Omega \sin \xi (R_s - r_1 \sin \phi \cos \theta) \\
& - \frac{1}{1-M_r} \frac{\partial M_r}{\partial t} (r_1 \cos \phi \cos \xi + r_1 \sin \xi \sin \phi \sin \theta)].
\end{aligned}$$

This will give the radiated sound from each of the blades. The above expression is to be evaluated at the retarded time. This will require some further manipulations.

$$t_R = t - r/a_0 \quad \text{where } r = |\vec{x} - \vec{y}|$$

square the distance r

$$\begin{aligned}
 r^2 &= (x_i - y_i)^2 = (r_1 \cos \phi)^2 + (r_1 \sin \phi - R_s \cos \theta)^2 + (R_s \sin \theta)^2 \\
 &= r_1^2 - 2 r_1 R_s \sin \phi \cos \theta + R_s^2 .
 \end{aligned}$$

Now, $r_1 \gg R_s$, so it will be sufficient for the far field solution to write

$$r^2 = r_1^2 - 2 r_1 R_s \sin \phi \cos \theta$$

or, equivalently,

$$r^2 = (r_1 - R_s \sin \phi \cos \theta)^2 - R_s^2 \sin^2 \phi \cos^2 \theta .$$

For the far field solution we can therefore use

$$r \approx r_1 - R_s \sin \phi \cos \theta .$$

The retarded time can be written as

$$t_R = t - r_1/a_o + R_s/a_o \sin \phi \cos [\theta] ,$$

where the $[\theta]$ is the retarded value of the blade location.

This source position may be written as

$$[\theta] = \Omega t_R + \theta_{t_R=0} = \Omega(t - r_1/a_o) + \frac{\Omega R_s}{a_o} \sin \phi \cos [\theta] + \theta_{t_R=0} .$$

Now in the solution for the radiated sound substitute t_R for t and $[\theta]$ for θ , the radiated sound pressure becomes

$$\begin{aligned}
 p_K(\vec{x}, t) &= \frac{F}{4\pi a_o r^2 (1 - [M_r])^2} \{ \Omega \sin \xi (R_s - r_1 \sin \phi \cos [\theta])_k \} \\
 &\quad - \frac{r_1}{(1 - [M_r])_k} \left[\frac{\partial M_r}{\partial t} \right]_k (\cos \xi \cos \phi + \sin \xi \sin \phi \sin [\theta]_k) \}
 \end{aligned}$$

where: $r = r_1 - R_s \sin \phi \cos [\theta]_k$

$$[M_r]_k = -M_s \sin \phi \sin [\theta]_k$$

$$[\partial M_r / \partial t]_k = -M_s \Omega \sin \phi \cos [\theta]_k ,$$

with: $[\theta]_k = \Omega(t - r_1/a_o) = (\Omega R_s/a_o) \sin \phi \cos [\theta]_k + \theta_{k,t_R=0}$

$\theta_{k,t=0}$ = location of kth source at $t_R = 0$.

The total sound from all the rotor blades becomes

$$p(\vec{x}, t) = \sum_{K=1}^B p_k(\vec{x}, t) .$$

4.2 How to Use Program TONE

As was seen in the previous section, the model used to predict the sound radiation generates a discrete time history using the solution for a rotating array of forces, then treats that history in the same way one would treat experimental data.

We previously mentioned the flexibility of this approach. This flexibility is achieved in the program by changing the program listing. A multiple option approach could have been taken but this is wasteful of computer time and assumes one knows all the options which will be needed. The reason for this approach was to provide a flexible tool; however, it does depend upon the operator's ability to change the Fortran statements used to generate the time series. This is easily done since the equations are short and the program is straightforward. The part of the program actually used to generate the time series is about 20 cards long.

The form of the program set up here is for the case where the force on each of the blades is the following function of time

$$F(t) = F_B (1 + C_d x(t))$$

where F_B is the steady state blade force, C_d is a dynamic loading coefficient, and $x(t)$ is a time series simulating the unsteady force fluctuations on the blade. Depending upon the particular mechanism, $x(t)$ can be chosen to simulate either wake related force fluctuations or the fluctuations caused by inflow turbulence interacting with the airfoil. Processes of this type can be represented using autoregressive processes. This will be discussed in section 4.5 below.

4.3 Program Inputs

Input to the program is done using cards. Four cards are required to run the program in its present form.

CARD 1, FORMAT(1I5)

NOPT	= 0, no plots
	= 1, plot the time history
	= 2, plot the sound spectrum
	= 3, plot both the history and the spectrum

NOPT controls the type of output desired. Two CALCOMP subroutines are contained in the program so that the time history for the process can be examined and/or the spectrum of the radiated sound can be plotted. If neither plot is desired then listings of the time history and the spectrum will be the only output.

CARD 2, FORMAT(7F10.2)

BE	bandwidth desired in the spectrum	(Hz)
FMAX	maximum frequency in the spectrum	(Hz)

See Section 4.4 for the implications of these selections.

CARD 3, FORMAT(7F10.2)

ZR	number of rotor blades	(--)
XI	blade stagger angle, measured from blade chord to radial plane passing through rotor axis	(DEG)
PHI	radiation angle, measured from inlet rotor axis	(DEG)
RPM	rotor speed	(RPM)
RS	radial distance to source	(M)
RI	radial distance to observer	(M)
F	magnitude of force on each blade	(KG)

CARD 4, FORMAT(7F10.2)

CDYN	dynamic force coefficient	(--)
A1	autoregression coefficient	(--)
A2	autoregression coefficient	(--)

See Section 4.5 for a discussion of the selection of the autoregression coefficients.

4.4 Time Sampling

The setting of FMAX and BE determines the sampling rate and the total number of required samples. Letting

$$f_{\max} = \text{FMAX}$$

$$B_e = \text{BE}$$

we can illustrate how this is done. First, estimate the number of samples required as

$$N = 2 f_{\max} / B_e.$$

Since we will use a Fast Fourier Transform to find the spectrum, the number of samples must be of the form $N' = 2^m$ where m is an integer. We choose m such that

$$2^{m-1} < N \leq 2^m.$$

Then compute $N' = 2^m$ samples of the time series. The new bandwidth becomes

$$B'_e = 2 f_{\max} / N'.$$

The sampling rate for the time series is

$$H = 1 / (2 f_{\max}).$$

4.5 Use of Autoregression Processes

A second order autoregression process is used in the current form of the program to represent the force fluctuations. This process may be written

$$x(t) = a_1 x(t-1) + a_2 x(t-2) + Z(t)$$

where a_1 and a_2 are the autoregression coefficients ($A1$ and $A2$ in the program). $Z(t)$ is a random number selected using a random number generator.

A time series is generated separately for each blade. If there is to be no correlation between different blades (as for wake related forces) then

Z will be taken from a different random series for each blade. If correlation is desired between blades then the time series can be selected to reflect the correlation. Correlation would be expected to exist between blades for the mechanism of inflow turbulence.

The main idea is to fit the time series to represent what is desired, then let the program compute the resulting sound radiation.

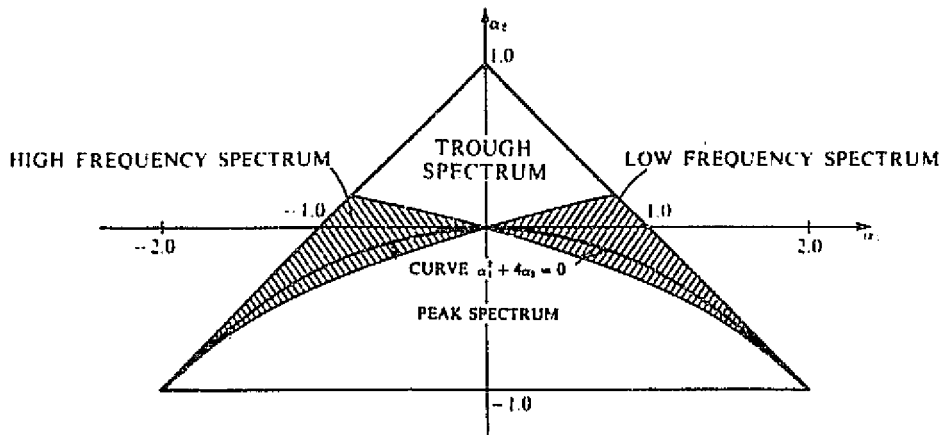
The spectrum for the second order process above can be solved for directly. The spectrum is

$$G_{xx}(f) = H\sigma_z^2 / (1 + a_1^2 + a_2^2 - 2a_1(1-a_2) \cos 3\pi fH - 2a_2 \cos 4\pi fH)$$

$$-1/2H \leq f \leq 1/2H.$$

Selection of a_1 and a_2 gives the spectral shape to be expected from $x(t)$.

A plot of a_1 and a_2 for different spectral shapes is shown below to give an idea of the values of the coefficients needed for particular shapes.



A peaked spectrum is what we would expect for a wake related sound generation mechanism. A low frequency spectrum would be expected for the cases of inflow turbulence. The frequency where the peak will occur is given by

$$\cos 2\pi f_0 H = -a_1 (1 - a_2)/4a_2 .$$

As may be expected, some experimentation will be required to fit the process to an actual experimental spectrum.

To obtain more complicated power spectral shapes higher order processes may be needed. These can be handled as easily as the above on the computer, but the preliminary efforts to get a particular spectrum increase.

For a good general discussion of the above method the reader is referred to SPECTRAL ANALYSIS AND ITS APPLICATIONS by G.M. Jenkins and D.G. Watts (1969).

If more simply definable fluctuations are needed these can be used for $x(t)$. For example, inlet distortion or the effect of another blade row can be simulated by a Fourier series representation of the force fluctuation resulting from the distortion. A simple cosine with period equal to the stator spacing will give a good qualitative model of blade row interaction noise.

The model can also be extended to account for span distributions of the blade force. This is straightforward but the computer time required does increase substantially.

4.6 Program Listing

```

PROGRAM TONE(INPUT,OUTPUT,TAPE5=INPUT,TAPE6=OUTPUT)

C   COMPUTE SOUND RADIATION FROM A ROTATING ARRAY
C   OF POINT FORCES

      COMPLEX C
      DIMENSION PRAD(1100),THEP(20),THE(20),THER(20),THERP(20)
      DIMENSION C(1024),B(2,1024),INV(256),M(3),S(256),AMP(1024),
1    ISPL(1024),FREQ(512),PRA(1024),Z(1024),X(4,1024),DATO(2048)
      EQUIVALENCE(C,B)

C   NHAR = HARMONICS TO BE ANALYZED
C   NOPI = OPTION
C   BE = FREQUENCY BANDWIDTH DESIRED
C   ZR = NUMBER OF ROTOR BLADES
C   XI = STAGGER ANGLE, MEASURED FROM FAN AXIS (DEG)
C   PHI = RADIATION ANGLE - MEASURED FROM FAN AXIS
C   RPM = ROTOR SPEED (RPM)
C   RS = SOURCE RADIUS (IN)
C   RI = OBSERVER RADIUS (IN)
C   F = FORCE ON EACH ROTOR BLADE (KG)
C   CDYN = DYNAMIC LOADING - PER CENT OF TOTAL FORCE
C   A1 = AUTO-REGRESSION CONSTANT
C   A2 = AUTO-REGRESSION CONSTANT

C       X(J,I) = A1*X(J,I-1) + A2*X(J,I-2) + Z(I)
C       Z(I) = RANDOM VARIABLE

      DATA PI,AV/3.14159,342./

C   INPUT M1 SUCH THAT C HAS LENGTH N=2**M1

      READ(5,1) NHAR,NOPI
1    FORMAT(2I5)
      READ(5,2) BE

C   OPTION = 0  NO PLOTS
C   OPTION = 1  PLOT TIME HISTORY
C   OPTION = 2  PLOT SPECTRUM
C   OPTION = 3  PLOT BOTH TIME HISTORY AND SPECTRUM

      READ(5,2) ZR,XI,PHI,RPM,RS,RI,F
      READ(5,2) CDYN,A1,A2
2    FORMAT(7F10.2)

C   CONDITION INPUT DATA

      CMFT = 3.28
      RS = RS*CMFT
      RI = RI*CMFT
      F = F*2.21
      FMAX=FLOAT(NHAR)*RPM/60.*ZR
      FMAX=10000.
      NI=IFIX(2.*FMAX/BF)
      DO 75 I=1,10

```

REPRODUCIBILITY OF THE
ORIGINAL PAGE IS POOR

```

      N2=2**I
      IF(N2.GT.N1) GO TO 71
75  CONTINUE
71  NSAMP=N2
      M1=I

```

C THE MAXIMUM NUMBER OF POINTS ARE 2**10 (1024)

```

      BE=2.*FMAX/FLOAT(NSAMP)
      XOMEG=RPM*2.*PI/60.
      XIP=XI*PI/180.
      PHIP=PHI*PI/180.
      SINPHI=SIN(PHIP)
      COSPHI=COS(PHIP)
      SINXI=SIN(XIP)
      COSXI=COS(XIP)
      XAS=RS*XOMEG/A0
      NZR=ZR

```

C RANDOM FORCE ON THE ROTOR BLADES MODELLED BY UNCORRELATED
C SECOND ORDER PROCESS

```

      IX=2**24+3
      NRAND=NSAMP
      DO 31 J=1,NZR
      IX=IX+10
      CALL RAND(10,IX,1,NRAND,4)
      X(J,1)=1.0
      X(J,2)=1.0
      DO 31 I=3,NSAMP
31  X(J,I)=A1*X(J,I-1)+A2*X(J,I-2)+Z(I)

```

C WRITE INPUT DATA

```

      WRITE(6,20) ZR,XI,RPM,F,RS,PHI,R1
20  FORMAT(1H,51H RADIATION OF SOUND FROM A ROTATING ARRAY OF FORCES
      1//20H NUMBER OF ROTOR BLADES 1F5.0/20H BLADE STAGGER ANGLE
      21F5.2/18H ROTOR SPEED (RPM) 1F10.2/12H BLADE FORCE 1F10.2/
      35H RADIAL LOCATION OF POINT FORCE 1F5.2/
      416H RADIATION ANGLE 1F5.2/10H OBSERVER RADIOS 1F5.1//)
      WRITE(6,22) FMAX,BE
22  FORMAT(//18H MAXIMUM FREQUENCY
      11F20.3,15H BANDWIDTH (HZ) 1F10.2//)
      WRITE(6,23) CDYN,A1,A2
23  FORMAT(27H RANDOM FORCE FLUCTUATIONS //
      137H X(I) = A1*X(I-1) + A2*X(I-2) + Z(I) /
      223H DYNAMIC COEFFICIENT = 1F10.3 /
      35H A1 = 1F10.3,10X,3H A2 = 1F10.3 //)

```

C INITIALIZE BLADE POSITIONS

```

      DO 10 I=1,NZR
      UI=I
10  THE(I)=2.*PI/ZR*(UI-1.)

```

C COMPUTE TIME SERIES BASED ON RETAINING SUFFICIENT
C INFORMATION FOR 10 HARMONICS

C 30, SAMPLE AT 20 SAMPLES PER BLADE SPACING

C SET TIME INCREMENT

FB=F
DT=0.5/FMAX
DO 200 K=1,NSAMP
JK=K
T=DT*(JK-1.)

C COMPUTE SOUND RADIATION FROM THE BLADE ARRAY

PRAD(K)=0.0
DO 300 J=1,NZR
JJ=J
IF(K.NE.1) GO TO 301
THERP(J)=THE(J)
GO TO 302
301 THERP(J)=THERP(J)
302 CONTINUE
THER(J)=XOMEG*(T-R1/A0+NS/A0*SINPHI*COS(THERP(J)))+THE(J)
THERP(J)=THER(J)
XMR=-XNS*SINPHI*SIN(THER(J))
JXMR=-XNS*XOMEG*SINPHI*COS(THER(J))
R=R1-RNS*SINPHI*COS(THER(J))
F=FB*(1.+CDYN*X(J,K))
VAR1=F/((4.*PI*A0*R**2)*(1.-XMR)**2)
VAR2=XOMEG*SINPHI*(NS-N1*SINPHI*COS(THER(J)))
VAR3=-R1*JXMR/(1.-XMR)*(COSX1*COSPHI+
1/SINX1*SINPHI*SIN(THER(J)))
PBLADF=VAR1*(VAR2+VAR3)
300 PRAD(K)=PRAD(K)+PBLADF
200 CONTINUE

C THIS COMPLETES COMPUTATION OF THE TIME SERIES FOR
C ONE RADIATION ANGLE

WRITE(6,21) (I,PRAD(I),I=1,NSAMP)
21 FORMAT(4(5X,115,5X,1E15.3))

USAMP=NSAMP

C FIND THE MEAN VALUE

PSUM=0.0
DO 400 I=1,NSAMP
400 PSUM=PSUM+PRAD(I)
PMEAN=PSUM/USAMP

WRITE(6,30) PMEAN
30 FORMAT(///29H MEAN RADIATED SOUND PRESSURE 1E20.5//)
DO 500 I=1,NSAMP
500 PRA(I)=PRAD(I)/PMEAN

IF(NOPT.EQ.1) GO TO 701
IF(NOPT.EQ.4) GO TO 701

GO TO 702
701 CONTINUE

C CONDITION THE DATA FOR THE TIME PLOT
C FIND MAXIMUM AND MINIMUM VALUES IN THE PRESSURE ARRAY

 PMAX=-999 & PMIN=999
 DO 600 I=1,NSAMP
 IF(PRA(I).LT.PMIN) PMIN=PRA(I)
 IF(PRA(I).GT.PMAX) PMAX=PRA(I)
600 CONTINUE

C PLOT THE NORMALIZED ACOUSTIC PRESSURES
 CALL IPLOT(PRA,NSAMP,DT,PMIN,PMAX,NZR)

702 CONTINUE

C PROVIDE ARRAY C
 DO 40 I=1,NSAMP
 II=2*I
40 DATA(II-1)=PRAD(I)
 NNN=2*NSAMP
 WRITE(6,41) (DATA(I),I=1,NNN)
 CALL FOUR1(DATA,NSAMP,-1)
 DO 70 I=1,NSAMP
 II=2*I
70 C(I)=CMPLX(DATA(II-1),DATA(II))
C ARRAY C NOW CONTAINS THE FREQUENCY COMPONENTS
C COMPUTE AMPLITUDE AND PHASE ANGLE FORM
 WRITE(6,41) (C(I),I=1,50)
41 FORMAT(2(10X,1E15.3,5X,1E15.3))
 NF=NSAMP/2
 DO 50 I=1,NF
 AMP(I)=CABS(C(I))
 AMP(I)=AMP(I)/NF
50 SPL(I)=20.*ALOG10(AMP(I)/20.*E-06)

C USE FAST FOURIER TRANSFORM TO EXPRESS IN TERMS OF FREQUENCY

C COMPUTE THE FREQUENCY ARRAY

 FMAX=1./(2.*DT)
 DF=FMAX*2./FLOAT(NSAMP)
 DO 60 I=1,NF
 UI=I
60 FREQ(I)=BF*UI

 WRITE(6,61)
61 FORMAT(//24H RADIATED SOUND SPECTRUM //)
 WRITE(6,62) (FREQ(I),SPL(I),I=1,NF)
62 FORMAT(4(1F10.2,5X,1F10.2,5X))


```

IF(NOPT.EQ.2) GO TO 703
IF(NOPT.EQ.4) GO TO 703
GO TO 1000

```

```

703 CONTINUE

```

```

C NOW PLOT THE SPECTRUM OF THE RADIATED SOUND
C CONDITION THE DATA FOR THE SPECTRAL PLOT

```

```

NF=NSAMP/2
CALL FPLOT(SPL,FREQ,NF)

```

```

1000 CONTINUE

```

```

STOP

```

```

END

```

```

SUBROUTINE FOUR1(DATA,N,ISIGN)

```

```

C A SIMPLE FAST FOURIER TRANSFORM

```

```

DIMENSION DATA(2048)
PI=ACOS(-1.)
TPI=2.*PI
IP0=2
IP3=IP0*N
I3REV=1
DO 50 I3=1,IP3,IP0
IF(I3-I3REV)10,20,20
10 TEMPR=DATA(I3)
TEMPI=DATA(I3+1)
DATA(I3)=DATA(I3REV)
DATA(I3+1)=DATA(I3REV+1)
DATA(I3REV)=TEMPR
DATA(I3REV+1)=TEMPI
20 IP1=IP3/2
30 IF(I3REV-IP1) 50,50,40
40 I3REV=I3REV-IP1
IP1=IP1/2
IF(IP1-IP0)50,30,30
50 I3REV=I3REV+IP1
IP1=IP0
60 IF(IP1-IP3) 70,100,100
70 IP2=IP1*2
THE1A=TPI/FLOAT(ISIGN*IP2/IP0)
SINH=sin(THE1A/2.)
COS1R=-2.*SINH*SINH
COS1I=sin(THE1A)
WR=1.
WI=0.
DO 90 I1=1,IP1,IP0
DO 80 I3=I1,IP3,IP2
I2A=I3
I2B=I2A+IP1
TEMPR=WR*DATA(I2B)-WI*DATA(I2B+1)
TEMPI=WR*DATA(I2B+1)+WI*DATA(I2B)
DATA(I2B)=DATA(I2A)-TEMPR
DATA(I2B+1)=DATA(I2A+1)-TEMPI
DATA(I2A)=DATA(I2A)+TEMPR
DATA(I2A+1)=DATA(I2A+1)+TEMPI
80
90

```

```

    TEMPR=WR
    WK=WR*WSTPR-WI*WSTPI+WK
90  WI=WI*WSTPR+TEMPR*WSTPI+WI
    IP1=IP2
    GO TO 60
100 RETURN
    END
    SUBROUTINE TPLOT(PRAD,NSAMP,DI,PMIN,PMAX,NPAS)
    DIMENSION PRAD(1100),T(1100)

C    INITIALIZE THE PLOT

    CALL PRNTON

C    PLOT SIZE TO BE 5 BY 7 INCHES
C    SET OBJECT SPACE

C    NPAS IS THE NUMBER OF BEAD PASSAGES TO BE PLOTTED

    UPAS = NPAS
    TMAX=UPAS*20.*DT

    CALL STS2UB(2.,9.,2.,7.)
    CALL STSUBJ(0.,TMAX,PMIN,PMAX)

C    CONSTRUCT AXES

    CALL STNDIV(1,2)
    CALL AXLIL1
    CALL SAXLIT
    CALL SAXLIR

C    LABEL SCALES

    CALL STNDIV(1,2)
    CALL STNDEC(0)
    CALL NOSLIB
    CALL NOSLIL

C    TITLE SCALES

    CALL STNCHR(11)
    CALL TITLED(11H TIME (SEC) )
    CALL STNCHR(29)
    CALL TITLED(29H FAR FIELD ACOUSTIC PRESSURE )

C    GENERATE THE TIME ARRAY

    NT=NPAS*20
    TI=0.0
    DO 10 I=1,NT
    UI=I
10  T(I)=TI+DI*(UI-1.)

C    PLOT THE DATA USING STRAIGHT LINES

```

```
CALL STNPIS(NI)
CALL SLLILI(T,PRAD)
```

```
C   TERMINATE PLOTTING
```

```
CALL EXITPL
RETURN
END
SUBROUTINE FPLOT(SPL,FREQ,NF)
DIMENSION SPL(1024),FREQ(1024)
```

```
C   INITIALIZE THE PLOT
```

```
CALL STCCON(48H 1000 TONE NOISE GENERATION SPECTRUM
```

```
C   PLOT SIZE TO BE 5 BY 7 INCHES
```

```
C   SET OBJECT SPACE
```

```
CALL STGZOB(2.,9.,2.,7.)
CALL STGOSB(10.,10000.,50.,150.)
```

```
C   CONSTRUCT AXES
```

```
CALL STNDIV(30,20)
CALL AXGLI
CALL JAXLST
CALL JAXLIR
```

```
C   LABEL SCALES
```

```
CALL STNDIV(3,10)
CALL STNDEC(0)
CALL NOSLIL
CALL NOLGB
```

```
C   TITLE SCALES
```

```
CALL STNCHR(15)
CALL TITLBB(15H FREQUENCY (Hz) )
CALL STNCHR(37)
CALL TITLCL(37H RADIATED SPL (dB RE 20 MICRO W/4**2) )
```

```
C   PLOT THE DATA USING STRAIGHT LINES
```

```
CALL STNPIS(NF)
CALL SLLGLI(FREQ,SPL)
```

```
C   TERMINATE PLOTTING
```

```
CALL EXITPL
RETURN
END
```

5.0 PROGRAM SDATA

Program SDATA is used to perform correlation and spectral analysis on discrete time series. The particular operations carried out on two given time series are:

- 1) Autocovariance function for each sample and cross covariance between the two samples
- 2) Autocorrelation function for each sample and cross correlation between the two samples
- 3) Autospectrum for each sample
- 4) Magnitude and phase representation of cross spectrum between the two samples
- 5) Squared coherency function

The analysis used as the basis for the program is that presented in SPECTRAL ANALYSIS AND ITS APPLICATIONS, by G.M. Jenkins and D.G. Watts. The method used is exactly that recommended in this reference.

Two input possibilities exist. The first is to input two time series of arbitrary length. For this case the program will compute all the above covariances, correlations, and spectral quantities. The second possibility is to input two covariance functions. In this case the program will only carry out the spectral analysis. Some electronic instruments have become available recently which quickly compute correlation functions. This program can be used in conjunction with these instruments to obtain the spectral representation of the data.

A classical Fourier transform was used in the program. For this reason long samples, in terms of correlation lags, will require substantial computer time. Therefore, a Fast Fourier Transform program should be used for large samples.

5.1 Outline of Computations

This outline is substantially the same as that presented by Jenkins and Watts (pages 382 and 383).

Two time series of data, x_{t1} and x_{t2} , sampled at increments of H sec for a total of N points. For convenience H is assumed equal to one. The frequency range is then $0 \leq f \leq 1/2$ Hz. For non-unity values of H the frequency range is $0 \leq f \leq 1/2H$. The actual spectral levels are also to be multiplied by H to obtain quantitatively correct values.

The computations are:

(1) For the x_{t1} data:

(a) the autocovariance estimate is

$$c_{11}(k) = \frac{1}{N} \sum_{t=1}^{N-k} (x_{1t} - \bar{x}_1)(x_{1t+k} - \bar{x}_1) \quad 0 \leq k \leq M-1$$

where M is the number of time lags in the estimate and the mean value is

$$\bar{x}_1 = \frac{1}{N} \sum_{t=1}^N x_{1t}$$

(b) the smoothed spectral estimate using a Tukey window

$$\bar{c}_{11}(j) = 2 \left\{ c_{11}(0) + 2 \sum_{k=1}^{M-1} c_{11}(k) w(k) \cos \frac{\pi k j}{F} \right\} \quad 0 \leq j \leq F$$

F is a measure of the spacing of the frequency points. The smoothed spectral estimates are to be computed at $0, 1/2F, \dots, 1/2$ where F is of the order 2 to 3 times M . The Tukey window $w(k)$ is defined by

$$w(k) = \frac{1}{2} \left(1 + \cos \frac{\pi k}{M} \right) \quad |k| \leq M$$

$$= 0 \quad |k| > M.$$

(2) For the x_{t2} data:

(a) the autocovariance function estimate

$$c_{22}(k) = \frac{1}{N} \sum_{t=1}^{N-k} (x_{2t} - \bar{x}_2)(x_{2t+k} - \bar{x}_2) \quad 0 \leq k \leq M-1$$

with the mean value

$$\bar{x}_2 = \frac{1}{N} \sum_{t=1}^N x_{2t}.$$

(b) the smoothed spectral estimate

$$\bar{c}_{22}(j) = 2 \left\{ c_{22}(0) + 2 \sum_{k=1}^{M-1} c_{22}(k) w(k) \cos \frac{\pi k j}{F} \right\} \quad 0 \leq j \leq F$$

where the same window is used (Tukey) for $w(k)$.

(3) For the x_{t1} and the x_{t2} data:

(a) the cross covariance estimate

$$c_{12}(k) = \frac{1}{N} \sum_{t=1}^{N-k} (x_{1t} - \bar{x}_1)(x_{2t+k} - \bar{x}_2) \quad 0 \leq k \leq M-1$$

$$c_{21}(k) = c_{12}(-k) = \frac{1}{N} \sum_{t=1}^{N-k} (x_{1t+k} - \bar{x}_1)(x_{2t} - \bar{x}_2) \quad 0 \leq k \leq M-1$$

(b) the even and odd cross covariance estimates

$$e_{12}(k) = \frac{1}{2} \{ c_{12}(k) + c_{12}(-k) \} \quad 0 \leq k \leq M-1$$

$$o_{12}(k) = \frac{1}{2} \{ c_{12}(k) - c_{12}(-k) \} \quad 0 \leq k \leq M-1$$

(c) the smoothed co- and quadrature spectral estimates

$$\begin{aligned}\bar{L}_{12}(j) &= 2 \{ \ell_{12}(0) + 2 \sum_{k=1}^{M-1} \ell_{12}(k) w(k) \cos \frac{\pi j k}{F} \} & 0 \leq j \leq F \\ \bar{Q}_{12}(j) &= 4 \sum_{k=1}^{M-1} q_{12}(k) w(k) \sin \frac{\pi j k}{F} & 1 \leq j \leq F-1\end{aligned}$$

note that $q_{12}(0) = 0$ and that $Q_{12}(0) = Q_{12}(F) = 0$.

(d) the smoothed cross amplitude spectral estimate

$$\bar{A}_{12}(j) = \sqrt{\bar{L}_{12}^2(j) + \bar{Q}_{12}^2(j)} \quad 0 \leq j \leq F$$

(e) the smoothed phase spectral estimate

$$\bar{\Phi}_{12}(j) = \arctan \left\{ - \frac{\bar{Q}_{12}(j)}{\bar{L}_{12}(j)} \right\} \quad 0 \leq j \leq F$$

(f) the smoothed squared coherency spectral estimate

$$\bar{k}_{12}^2(j) = \frac{\bar{A}_{12}^2(j)}{\bar{c}_{11}(j) \bar{c}_{22}(j)} \quad 0 \leq j \leq F$$

5.2 Correction for Bias Error Reduction

When there is a large delay in the correlation a large error can result in the coherency and phase spectrum. This bias error can be reduced by shifting the cross covariance function so that the maximum correlation occurs at zero time lag.

The way this correction is made in the present program is to search for the time lag corresponding to the largest absolute value of covariance. A new covariance function is then generated such that the peak occurs at the

zero time lag. This function is Fourier transformed and the phase shift introduced by the time shift is applied to the spectral estimate to correct it back. A consequence of the generation of a new covariance function is that it necessarily will contain fewer terms than the original. This results in a wider frequency bandwidth in the spectral estimate. This is why the program will seem to generate different bandwidths for different batches of data even when the same number of lags are called out in the input.

5.3 Input to Program

Two possibilities exist for input to this program. The first is to read in two time series of variables for which correlation and spectral computations are required. This is obtained from data card one by setting the variable `OPTION = 1, FORMAT(1I5)`. If `OPTION` is not set equal to one the program will read in the data as if it were covariance functions. Input for `OPTION = 1` is described in Section 5.3.1 and for `OPTION \neq 1` in Section 5.3.2.

CARB 1, FORMAT(1I5)

```
OPTION      = 1 read in time series
            ≠ 1 read in covariance functions
```

5.3.1 OPTION = 1

CARD 2, FORMAT(2I5)

NSAMP	number of points in the time sample
MLAG	number of time lags desired in the covariance functions

typically MLAG is chosen to be approximately one tenth as large as NSAMP.

CARD 3, FORMAT(1F10.2)

DELTA time increment in samples (Sec)

The remaining inputs under this option are the actual values of the variable in the time series. They are read 7 values to each card in the FORMAT(7F10.2). The number of cards required depends upon the size of the sample. The computer call is

```
DO 10 I = 1,2
10 READ(5,2) (X(J,I), I = 1, NSAMP)
```

This is the total input required for this option.

5.3.2 OPTION # 1

CARD 2, FORMAT(2F10.2)

XLAG	number of values in the covariance function samples	
DELT	time increment in the functions	(Sec)

The remaining cards are used to read in the covariance functions. In the present format the functions are read in as follows.

```
DO 10 I = 1,2
DO 10 J = 1,2
10 READ(5,2) (COV(K,I,J), K = 1,M1)
```

where $M1 = XLAG + 1$ and the indices I, J mean the following:

$COV(K,1,1) = C_{11}(k)$ = autocovariance function for series 1, $k = 0, M$
 $COV(K,2,2) = C_{22}(k)$ = autocovariance function for series 2, $k = 0, M$
 $COV(K,1,2) = C_{12}(k)$ = cross covariance function for positive time lags, $k = 0, M$
 $COV(K,2,1) = C_{21}(k)$ = cross covariance function for negative time lags, $k = 0, M$

The format used to read in these functions is FORMAT(7F10.2). This completes the input requirements for this option.

5.4 Program Input

The output of the present form of this program is simply a listing of each of the functions computed. The list of functions computed was presented in Section 1.0.

```

PROGRAM SDATA(INPUT,OUTPUT,TAPE5=INPUT,TAPE6=OUTPUT)
C  A STATISTICAL DATA ANALYSIS PROGRAM WHICH COMPUTES THE FUNCTIONS
C  AUTO COVARIANCE AND CROSS COVARIANCE AS WELL AS
C  AUTO AND CROSS SPECTRUM
  DIMENSION X(1024,2),COV(101,2,2),Z(200)

C  READ IN THE TIME SERIES

C  NSAMP = NUMBER OF TIME SAMPLES
C  MLAG = NUMBER OF CORRELATION LAGS
C  DELT = TIME INCREMENT (SEC)
C  X(I,J) = THE TWO SAMPLES TO BE ANALYZED

  DATA PI/3.14159/
  READ(5,1) OPTION

C  OPTION = 1 READ TIME SERIES OTHERWISE READ COVARIANCES

  IF(OPTION.EQ.1) GO TO 50
  READ(5,1) NSAMP,MLAG
  READ(5,2) DELT
  DO 10 I=1,2
10  READ(5,2) (X(J,I),J=1,NSAMP)
  WRITE(6,40) (I,X(I,1),X(I,2),I=1,NSAMP)
40  FORMAT(2(5X,1I5,5X,2E20.5))

C  FORM COVARIANCES AND CORRELATIONS

  CALL COVAR(NSAMP,X,COV,MLAG)

C  COMPUTE THE CROSS POWER SPECTRAL DENSITY

  GO TO 22
50  READ(5,2) XLAG,DELT
  MLAG=IFIX(XLAG)
  M1=MLAG+1
  DO 21 J=1,2
  DO 21 I=1,2
21  READ(5,2) (COV(K,I,J),K=1,M1)
22  CONTINUE

C  COVARIANCES ARE READ IN THE ORDER (1,1),(1,2),(2,1),(2,2)

  CALL CRSPEC(COV,MLAG,DELT,NSAMP)
1  FORMAT(2I5)
2  FORMAT(7F10.2)
  STOP
  END
  SUBROUTINE COVAR(NSAMP,X,COV,MLAG)
  DIMENSION SM(2),X(1024,2),COV(101,2,2),DCOV(101,2,2),
1COK(101,2,2),DCOK(101,2,2),SM(2),XM(2)

C  THE AUTO AND CROSS COVARIANCES AS WELL AS THE AUTO AND
C  CROSS CORRELATIONS ARE COMPUTED
C  COMPUTE MEAN VALUES

```

```

DO 1 J=1,2
SUM(J)=0.0
DO 2 I=1,NSAMP
2 SUM(J)=SUM(J)+X(I,J)
1 XM(J)=SUM(J)/NSAMP

```

C PRINT MEAN VALUES

```

WRITE(6,10)
10 FORMAT(1H+,36H RESULTS OF CORRELATION COMPUTATIONS )
WRITE(6,11) (XM(J),J=1,2)
11 FORMAT(//24H MEAN VALUE SERIES ONE = 1E20.5 /
124H MEAN VALUE SERIES TWO = 1E20.5 //)

```

C COMPUTE DEVIATIONS FROM THE MEAN

```

DO 3 J=1,2
DO 3 I=1,NSAMP
3 A(I,J)=X(I,J)-XM(J)

```

C COMPUTE COVARIANCES

```

M1=MLAG
DO 4 K=1,M1
DO 4 J=1,2
DO 4 L=1,2
SUM1=0.0
M2=NSAMP-K
DO 5 I=1,M2
5 SUM1=X(I,J)*X(I+K-1,L)+SUM1
4 COV(K,J,L)=SUM1/FLOAT(NSAMP)

```

C COMPUTE DIFFERENCED COVARIANCES

```

DO 6 L=1,2
DO 6 J=1,2
DCOV(1,J,L)=-COV(2,L,J)+2.*COV(1,J,L)-COV(2,J,L)
M3=M1-1
DO 6 K=2,M3
6 DCOV(K,J,L)=-COV(K-1,J,L)+2.*COV(K,J,L)-COV(K+1,J,L)
WRITE(6,12)
12 FORMAT(31H TABLE OF AUTO COVARIANCE (1-1) )
WRITE(6,14) (K,COV(K,1,1),K=1,M1)
14 FORMAT(4(1I5,5X,1F15.5,5A))
WRITE(6,15)
15 FORMAT(31H TABLE OF AUTO COVARIANCE (2-2) )
WRITE(6,14) (K,COV(K,2,2),K=1,M1)
WRITE(6,16)
16 FORMAT(32H TABLE OF CROSS COVARIANCE (1-2) )
WRITE(6,14) (K,COV(K,1,2),K=1,M1)
WRITE(6,17)
17 FORMAT(32H TABLE OF CROSS COVARIANCE (2-1) )
WRITE(6,14) (K,COV(K,2,1),K=1,M1)

```

C OUTPUT DIFFERENCED COVARIANCES

```

WRITE(6,18)
18 FORMAT(1H1,46H DIFFERENCED COVARIANCES (SAME ORDER AS ABOVE)  //)

C   COMPUTE CORRELATIONS NORMALIZED COVARIANCES
C   BOTH NORMAL AND DIFFERENCED FORMS

DO 20 L=1,2
DO 20 J=1,2
COVM=SQRT(COV(1,J,J)*COV(1,L,L))
DCOVM=SQRT(DCOV(1,J,J)*DCOV(1,L,L))
DO 20 K=1,M1
COR(K,J,L)=COV(K,J,L)/COVM
20 DCOR(K,J,L)=DCOV(K,J,L)/DCOVM

C   PRINT THE CORRELATIONS AND THEIR DIFFERENCED FORMS

WRITE(6,21)
21 FORMAT(1H1,50H AUTO AND CROSS CORRELATIONS AND DIFFERENCED FORMS
1///48H AUTO CORRELATION (WITH DIFFERENCED FORM) (1-1)  //)
WRITE(6,14) (K,COR(K,1,1),K=1,M1)
WRITE(6,22)
22 FORMAT(///43H AUTO CORRELATION (2-2) )
WRITE(6,14) (K,COR(K,2,2),K=1,M1)
WRITE(6,23)
23 FORMAT(///25H CROSS CORRELATION (1-2) )
WRITE(6,14) (K,COR(K,1,2),K=1,M1)
WRITE(6,24)
24 FORMAT(///25H CROSS CORRELATION (2-1) )
WRITE(6,14) (K,COR(K,2,1),K=1,M1)

C   THIS COMPLETES THE CORRELATION PHASE

RETURN
END
SUBROUTINE CROPEC(COV,MLAG,DEL1,NSAMP)

C   THE CROSS POWER SPECTRAL DENSITY IS COMPUTED

DIMENSION COV(101,2,2),COVP(101),SPECP(303),SPEC(303,2),FREQ(303),
1EV(101),XOD(101),SQ(303),COSPEC(303),ASPEC(303),PHASE(303),
2COHSQ(303),ASPEC(303,2),DUM(202),DUP(202)
DIMENSION CSPEC(303),FREC(303)
DATA PI/3.14159/
M1=MLAG

C   FIND S WHERE S IS THE NUMBER OF TIME LAGS TO ALIGN THE TWO
C   PROCESSES WITH MAXIMUM CORRELATION AT ZERO
C   DETERMINE LAG TIME FOR SERIES ALIGNMENT

WRITE(6,70) (COV(I,1,2),COV(I,2,1),I=1,M1)
DO 21 K=1,MLAG
21 DUM(K)=COV(M1+1-K,2,1)
DO 22 K=2,M1
22 DUM(MLAG+K-1)=COV(K,1,2)
MM=2*MLAG-1
WRITE(6,26) (I,DUM(I),I=1,MM)

```

REPRODUCIBILITY OF THE
ORIGINAL PAGE IS POOR

```

26 FORMAT(4(5X,1I5,5X,1E15.5))
   XMAX=-999.
   DO 23 K=1,MM
     DUP(K)=ABS(DUM(K))
     IF(DUP(K).GT.XMAX) GO TO 24
   GO TO 23
24 XMAX=DUP(K)
   IS=K-M1
23 CONTINUE
   WRITE(6,25) IS
25 FORMAT(///22H ALIGNMENT LAG NUMBER    1110///)

```

```

C    CONDITION THE DATA FOR THE CALCULATION OF THE
C    CROSS SPECTRUM

```

```

M2=M1-IABS(IS)

```

```

C    CONDITION THE DATA FOR AUTOSPECTRUM CALCULATION

```

```

DO 20 K=1,2
DO 10 I=1,M1
10 COVP(I)=COV(I,K,K)
   CALL AUSPEC(MLAG,DELT,COVP,SPEC,FREQ,IS)

```

```

C    STORE AUTOSPECTRUM

```

```

NF=2*M2+1
DO 11 I=1,NF
11 SPEC(I,K)=SPEC(I)
20 CONTINUE
   EV(I)=DUM(M1+IS)
   XOD(I)=0.0
   DO 40 K=2,M2
     NMM=M1+IS+K-1
     NMMM=M1+IS-K+1
     EV(K)=0.5*(DUM(NMM)+DUM(NMMM))
40 XOD(K)=0.5*(DUM(NMM)-DUM(NMMM))
   WRITE(6,70) (EV(I),XOD(I),I=1,M2)

```

```

C    DO THE FOURIER TRANSFORM

```

```

CALL CROSPEC(EV,XOD,COSPEC,QSPEC,MLAG,DELT,IS)

```

```

C    COMPUTE PHASE ANGLE AND COHERENCE FUNCTION

```

```

C    RETURN CROSS SPECTRUM TO UNSHIFTED FORM

```

```

DO 60 I=1,NF
  ARG=PI*FLOAT(I-1)*FLOAT(IS)/FLOAT(NF)
  CN=COS(ARG)
  SN=SIN(ARG)
  DUM1=COSPEC(I)*SN+QSPEC(I)*CN
  DUM2=COSPEC(I)*CN-QSPEC(I)*SN
  COSPEC(I)=DUM2
  QSPEC(I)=DUM1
60

```

```

SQ(I)=COSPEC(I)*COSPEC(I)+QSPEC(I)*QSPEC(I)
IF(COSPEC(I).EQ.0.0) GO TO 80
PHASE(I)=ATAN2(-QSPEC(I),COSPEC(I))
GO TO 81
80 PHASE(I)=PI/2.
81 CONTINUE
PHASE(I)=PHASE(I)*180./PI
CSPEC(I)=SQRT(SQ(I))
COHSQ(I)=SQ(I)/(SPEC(I,1)*SPEC(I,2))
IF(SPEC(I,1).LE.0.0) GO TO 51
GO TO 55
52 CONTINUE
IF(SPEC(I,2).LE.0.0) GO TO 53
GO TO 54
51 ASPEC(I,1)=-100.
GO TO 52
53 ASPEC(I,2)=-100.
GO TO 60
55 CONTINUE
ASPEC(I,1)=ALOG10(SPEC(I,1))
GO TO 52
54 CONTINUE
ASPEC(I,2)=ALOG10(SPEC(I,2))
60 CONTINUE

```

C GENERATE THE FREQUENCY ARRAY

```

BL=1.33/(FLOAT(MLAG)*DEL1)
DO 41 I=1,NF
41 FREQ(I)=FLOAT(I-1)/(2.*DEL1*FLOAT(NF-1))

```

C WRITE THE SPECTRAL LEVELS (CROSS PSD), PHASE AND COHERENCY

```

WRITE(6,61)
61 FORMAT(1H1,35H RESULTS OF POWER SPECTRAL ANALYSIS   ///
119H AUTOSPECTRUM (1-1)   ///)
70 FORMAT(3(5X,1F10.4,5X,1E15.5))
WRITE(6,70) (FREQ(I),SPEC(I,1),I=1,NF)
WRITE(6,62)
62 FORMAT(///19H AUTOSPECTRUM (2-2)   ///)
WRITE(6,70) (FREQ(I),SPEC(I,2),I=1,NF)
WRITE(6,63)
63 FORMAT(///15H CROSS SPECTRUM   ///)
WRITE(6,70) (FREQ(I),CSPEC(I),I=1,NF)
WRITE(6,64)
64 FORMAT(///12H PHASE ANGLE   ///)
WRITE(6,70) (FREQ(I),PHASE(I),I=1,NF)
WRITE(6,65)
65 FORMAT(///11H COHERENCY   ///)
WRITE(6,70) (FREQ(I),COHSQ(I),I=1,NF)
RETURN
END
SUBROUTINE CROSPEC(EV,XCD,COSPEC,QSPEC,MLAG,DEL1,IS)
DIMENSION EV(101),XCD(101),COSPEC(303),QSPEC(303),W(101)

```

C COMPUTES SMOOTHED DO- AND QUAD- SPECTRA AND SQUAREDAMPLITUDE

```

DATA PI/3.14159/
M1=MLAG
M=M1-IABS(IS)
M2=M-1
NF=2*M+1

```

C COMPUTE WEIGHTS USING TUKEY WINDOW

```

DO 20 J=1,M2
20 W(J)=0.5*(1.+COS(PI*FLOAT(J)/FLOAT(M)))
DO 10 I=1,NF
SUM1=0.0
SUM2=0.0
DO 30 K=1,M2
ARG=PI*FLOAT(K)*FLOAT(I-1)/FLOAT(NF-1)
SN=SIN(ARG)
CS=COS(ARG)
SUM1=SUM1+EV(K+1)*W(K)*CS
30 SUM2=SUM2+XOD(K+1)*W(K)*SN
COSPEC(I)=2.*DELT*(EV(1)+2.*SUM1)
10 USPEC(I)=4.*DELT*SUM2
RETURN
END
SUBROUTINE AUSPEC(MLAG,DELT,COV,SPEC,FREQ,IS)

```

C COMPUTE AUTOSPECTRUM OF COVARIANCES

```

DIMENSION COV(101),W(101),SPEC(303),FREQ(303)
DATA PI/3.14159/

```

C THE TUKEY WINDOW IS USED FOR DATA SMOOTHING
C COMPUTE WEIGHTS

```

M1=MLAG
M=M1-IABS(IS)
M2=M-1
NF=2*M+1
DO 10 I=1,NF
10 W(I)=0.5*(1.+COS(PI*FLOAT(I)/FLOAT(M)))

```

C CALCULATE A SMOOTHED AUTOSPECTRAL ESTIMATE

```

MM=M-1
DO 20 I=1,NF
SUM=0.0
DO 21 K=1,MM
ARG=PI*FLOAT(K)*FLOAT(I-1)/FLOAT(NF-1)
VAR=COV(K+1)*W(K)*COS(ARG)
21 SUM=SUM+VAR
20 SPEC(I)=2.*DELT*(COV(1)+2.*SUM)
C COMPUTE BANDWIDTH AND DEGREES OF FREEDOM
RETURN
END

```